### EXPERIMENTAL TELEVISION CENTER STUDIO DOCUMENTATION ( 2005 )

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2. Types of Video Sources and Methods of Processing

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- Sample Patch: Audio Patch Bays
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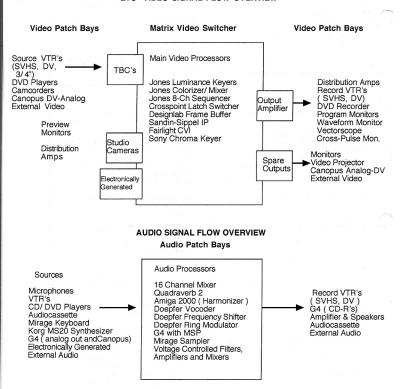
Section 2: + / -5V System Manual : Yellow Boxes ( Richard Brewster ) 1978 & 1984.

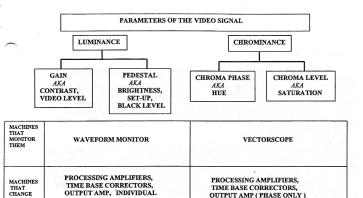
Section 3: Raster Manipulation Unit ( Sherry Miller Hocking w/ Richard Brewster and Walter Wright) 1978-80

TYPES OF SIGNALS IN THE S	YSTEM	METHODS OF ROUTING SIGNALS					
1. Composite Video 2. Y/C Video 3. Audio 4. +/-5 Volt 5. MIDI (Musical Instrument Di 6. Sync 7. DV 8. +/- 1/2 Volt (Sandin-Sippel I		Matrix Switcher and Video Patch Bay Black Svideo switch boxes / prepatched Audio Patch Bay Mono mini cables directly to and from devices MIDI Patch Bay Pre-patched Firewire cables directly to and from devices BNC cables directly to and from devices.					
TYPES OF AUDIO SOURCES  1. Microphones 2. Audiotape 3. Audio tracks of video media 4. CD Player 5. Electronically Generated Analog: Oscillators, Randon Sampled Sounds & Audio Sc		ETC STUDIO  Cardioid (2) Audio cassette deck (1) SVHS, DV, DVD G4, CD/ DVD Player  + /- 5 V System and Korg MS-20 Synthesizer Mirage Keyboard Sampler, Amiga 2000: Audiomaster, Harmonizer, M					
METHODS OF AUDIO PROCE	SSING:	G4: Max/MSP, Logic Express, FCP					
REAL TIME PROCESSING: CHANGING PARAMETERS O 1. Changing Waveshape  2. Changing Amplitude 3. Digital Processing: Harmonizer, Delay, Reverb Flanging,, Chorusing, Pitch 4. Switching & Mixing  SAMPLING, LOOPING & EDITE	, Ring Modulation, Shifting and Detuning.	Alesis Quadraverb 2 ( Q 2 ), +/- 5V: VC Filter, Waveshaper, Comparators / Doepfer: Vocoder, Frequency Shifter / Korg MS20 Filters,  +/- 5V: Dual VC Amplifier / Korg MS20 VCA / Doepfer: VC Mixer & Ring Modulator  Amiga: Harmonizer / Alesis Quadraverb 2 (Q2) G4: Max/MSP, Pluggo  8 Channel Sequencer ( w/ VC inputs and +/- 5 V output)  Behringer 16-Channel Mixer/ Doepfer VC Mixer/ +/- 5V: Dual VCA, Mixers  Amiga: Audiomaster, Mirage Keyboard Sampler,					
		G4 : Max/MSP, Logic Express, FCP					
TYPES OF CONNECTORS USI Connector BNC	ED IN THE SYSTEM Signals Composite Video	Interface in the ETC System Sandin-Sippel Image Processing System ( SSIP ) inputs and outputs of external video devices					
Trompeter RCA	Composite Video Composite Video/ Audio	Video Patch Bays External audio/video devices ( CD players, VHS decks )					
S-Video	Y/C Video	Switch Boxes to TBC's A, C and D, from camcorders,					
1/4" Mono	Audio	SWHE & DV decks, G4 analog output, external devices Audio Patch Bays, Behringer Mixer, Korg MS20 Synthesizer and external audio devices (keyboards, guitars, effects processors).					
Mini ( mono )	Control Voltages	+/ - 5 Volt System, Doepfer analog modules					
Mini ( stereo )	Audio	Headphone jacks, line outputs of portable audio devices (minidisc recorders, portable CD) audio to and from G4					
XLR Firewire	Audio DV	Microphones to Behringer Mixer mic inputs G4, DV decks & camcorders, DVD Recorder. Canopus ADVC-100					

TYPES OF VIDEO IMAGE SOURCES  1. Studio Cameras 2. Videotape 3. Electronically Generated Computer Generated Analog Generated Waveform Color  METHODS OF VIDEO IMAGE PROCESSING:	ETC STUDIO  Black and White (2), Color (4)  VHS, DV, 3/4" with Time Base Correctors  Amiga 2000: Dpaint IV, Brilliance  G4: Flash MX, After Effects, Text (FCP)  +/-5 Volt Analog System: Oscillators, Random Signal Generators  Jones Colorizer, Sandin-Sippel Image Processor
REAL TIME PROCESSING: CHANGING PARAMETERS OF A SINGLE IMAGE 1. Changing the gain and pedestal, chroma level, and phase 2. Changing polarity (image negative) 3. Removing chrominance (monochromatic) 4. Resizing, cropping and/or repositioning image 5. Real time digitization; changing image resolution 6. Strobing 7. Inverting image horizontally and vertically 8. Raster Deflection 9. Adding chrominance (colorization) 10. Manipulation of grey levels COMBINING IMAGES 1. Switching 2. Mixing	Time Base Correctors (TBC), Output Amp, Individual Channels of Keyers, Colorizer, and Sequencer Colorizer, Microtime DVE, Sandin-Sippel Image Processor (SSIP): Adder/Multiplier, Fairlight CVI Color Kills, TBC's  DVE, Fairlight  A-Live 2000, Designlab Frame Buffer (FB-1), DVE, Fairlight DVE, Fairlight  DVE, TBC A & C, FB-1, Fairlight  DVE, Paik-Abe Raster Device (aka Wobulator), Fairlight Wobulator  Jones Colorizer, FB-1, SSIP, Fairlight  SSIP: Amplitude Classifier, Differentiator, Function Generator  Crosspoint Latch Switcher Jones 8 Channel Sequencer  CL Switcher, SSIP: Adder/Multiplier and 36-Channel Mixer, Fairlight Jones Colorizer Jones Keyers 1 & 2, Individual Channels of Jones Colorizer, FB-1, SSIP: Adder/Multiple, Fairlight CL Switcher, Jones Keyers and Sequencer with +/- 5 V signals, Fairlight  TBC's, DVE, FB-1, Fairlight  TBC's, DVE, FB-1, Fairlight  G4 with Final Cut Pro (FCP), FB-1, Live 2000
Changing speed of direction of frame sequences	• G4 w/ FCP, FB-1, VHS # 3

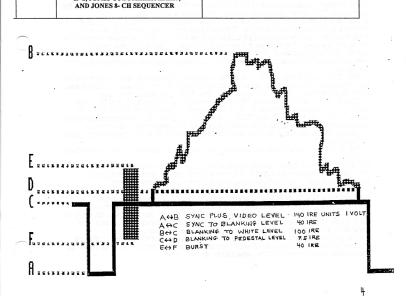
### ETC VIDEO SIGNAL FLOW OVERVIEW





CHANNELS OF JONES KEYERS 1 & 2, JONES COLORIZER/MIXER.

THEM



### Time Base Correctors:

Time Base Correctors, or TBC's are necessary for having sync information from an

external video signal be compatable with the rest of the ETC system.

Examples of external video signals include those coming from prerecorded tapes, a camcorder being used as a live camera in the studio, a digital video mixer (such as those from Videonics and Panasonic ) that are not otherwise externally syncable, and the Canopus ADVC-100 for converting a firewire signal from the Macintosh G4 to analog video. For any of these to be used as a source to the ETC system, they each have to pass through an individual TBC.

There are 5 TBC's in the system:

TBC A is a DPS Personal TBC III, a card inside the Amiga 4000 computer. It has both an svideo (Y/C) and composite video input. The composite video input is on the patch bay. The svideo input can be selected by the black switch box near the Amiga with a cable for external sources such as your camcorder. To switch between composite and s-video , you need to select the input in the TBC III software on the Amiga. The software also provides processing amplifier (or proc amp) controls for the TBC. The parameters are labeled "black" (pedestal), "video" (gain), hue (chroma phase) and "chroma" ( saturation) and are slider bar controls. There are also controls for freeze-field, freeze-frame and strobe. These are selected by a 3 position button on the menu. The "Take" button. when selected, will acttivate one of the three choices. There is a slider bar for controlling the strobe rate. The "Mono" button will make the image black and white.

TBC B is a For-A. It has a composite input only. The proc amp controls are a set of four switches and knobs on the front panel. For each parameter, when the switch is down, the knob is inactive and the signal output is at unity. When the switch is up, adjustments can be made with the knob. The parameters are labelled "video" (gain), "chroma level" ( saturation), "chroma phase", and "set-up" (pedestal). There are two red pushbuttons for freeze-field and freeze-frame. The "monochromatic" switch will make the video signal black

and white.

TBC C is a DPS Personal TBC IV, a card inside the Amiga 2000, and has both syideo and composite video inputs. It has an svideo switch box near the computer. The software

controls are similiar to TBC A. If you're using Jitter, TBC C is used as an output.

TBC D is also a DPS Personal TBC IV, card inside the Amiga 2000, and has both svideo and composite video inputs. The TBC IV menu controls both TBC's C and D. Select "TBC 1" to control C, and "TBC 2" to control D. Each TBC has seperate proc amp adjustments, as well as controls for strobe, freeze frame, monochrome, and svideo/ composite video selection. To choose an input type, select "PREFS", then select "svideo" or "composite"

TBC E is a For-A and has a composite video input only . Its controls are identical to those of TBC B

For all of the TBC's, the composite inputs are selected on the video patch bays. The TBC's have individual outputs on the video patch bay for sending the signal to

preview monitors. They also have three outputs each on the matrix switcher.

Every time a color signal is split on the matrix a certain amount of saturation is lost. If you are sending a TBC to more than one device in the sytem, use its multiple outputs on the matrix when available. For example, if you are sending TBC A to the Sequencer, the Crosspoint Latch Switcher and Jones Keyer 1 simultaneously, send matrix output H to the Sequencer, I to the switcher, and J to the keyer. There are also individual chroma level controls on four of the tbc's to boost or lower the saturation of that signal.

Although the overall hue of the final signal is determined by the Output Amp's chroma phase control, the relative hues of each tape source can be controlled to some extent by

the chroma phase controls on the individual tbc's.

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Sandin Sippel IP 3	4	4	+	+	1	1	1	H	1	1	1	1	L	F	F	F	Г	П	┚	I	I	Г	Г		I	I	T	Ι	T	1	
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Sandin Sippel IP 1	$\perp$	1	L	_	L	$\sqcup$		Ш	$\perp$	$\perp$	$\perp$	L	L	L	L	L			$\perp$	$\perp$	$\perp$	L	L			I	I	3	Ι	1	3 %
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# Experimental TV Center Video Patch Bays

#3 Amp Matrix Matrix DVD TBC W Record E Spare outputs of tbc's for sending to spare inputs to matrix Dist Amp monitors or to record decks. Amp Dist Dist Player Amp DVD tsiO Amp cameras, keyers, ) to preview monitors (or to Jitter via the Canopus) B&W monitors for preview Spare outputs from the matrix for sending individual devices (e.g. TBC ۵ Mon Mon Mon Mon Amp Amp 8 B B Mon Mon checking edges of the raster 8 cross-pulse monitor for Amp Dist Amp MiniDV DV Matrix Matrix Dist Amp \$20 #19 firewire for inputs analog signals to before going into the main system For converting corder g Source s must be sent to tbc's to the G4 ADVC VHS Cansndo Canopus ADVC VHS ¥ 뫉

Z

Black

TBC

Burst

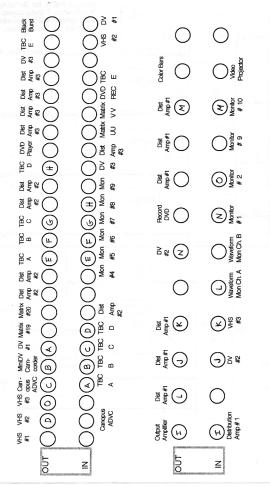
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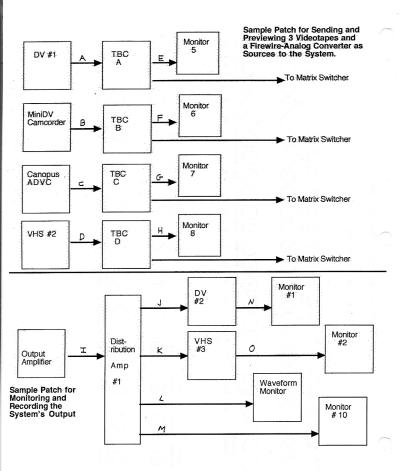
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Projector Color Bars Nideo Amp#1 vonitor Studio area monitors Amp#1 Monitor # 10 Amp#1 Vonitor Main program monitors Recorder Monitor DVD Mon Ch. B Waveform Also goes to the vectorscope for measuring both luminance and chrominance information Mon Ch. A Naveform video signal from the Panasonic DV deck must have its input select set for " A 1 " to record an analog Dist Amp #1 옷 patch bay. Amp #1 Amp #1 send it to several split a signal and Dist. Allows you to devices ( e.g. monitors and outpints are vtr's.) Six system for recording Distribution Main output of the Amp#1 Amplifier Output Z 7

## Experimental TV Center Video Patch Bays

### SAMPLE VIDEO PATCH



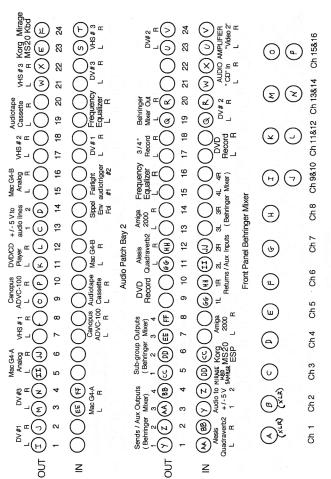


## Experimental Television Center Au. J Patch Bays 5-07-05 Audio Patch Bay 1

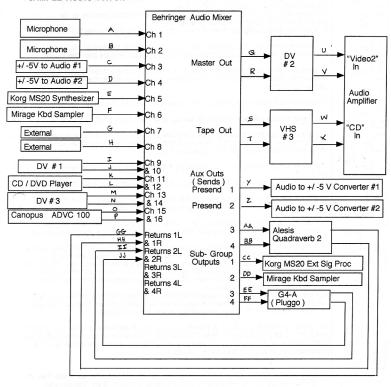
DV #1   DV #3   Analog   VHS#1   AUVC-100	DV#1 DV#3 Analog VHS#1 ADVC-  L R L R L R L R L R L L L L L L R L R	Mac G4-A	Sub-group Outputs DV (Behinger Miker) Recc	Alesis Audio to MAKE Korg Ampa 11.  Cuadraventz +1-5 v swipz MS20 2000 Re
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### SAMPLE AUDIO PATCH

### Audio Patch Bay 1

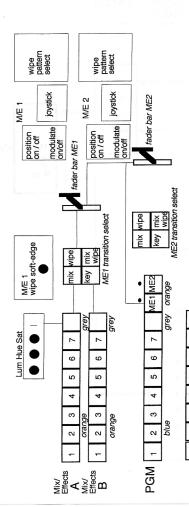


### SAMPLE AUDIO PATCH



Single Channel of	Behringer UB2442 Mixer:
Behringer Mixer	Connect audio sources directly to the mixer's inputs ( channels 1- 16) on the Behringer UB2442's front panel. Sources can be :
Gain Eq Hi	1. from the outputs of the patch bay (e.g. DV #1, DVD/CD Player, G4 analog audio) into the mixer's line inputs using 1/4" cables.     2. from external devices (e.g. your miniDV camcorder, minidisc recorder, guitar) into the mixer's line inputs using 1/4" cables or adaptors.     3. from microphones into the mixer's mic inputs (channels 1-12) using XLR connectors.
Mid	For microphone inputs, the channel's gain trimpots should be turned fully clockwise. For line inputs, the channel's gain trimpopts start straight up, and are then fine-tuned.
Freq	To send an individual channel to the main outputs, check that the "Main" switch is down ( on ) for that channel, and that the main mix fader bar is up as well as the indivudual fader bar for that channel.
U)Lo	The Behringer's main mix outputs are accessible from the patch bay.
Aux	Solo and mute switches should all be up and their LED's off.
2	Using the <b>aux sends</b> and <b>sub-groups</b> ,the mixer also acts as an extension of the audio routing system , allowing you to split signals and send them to several destination points (e.g. the +/-5V system, the G4, Quadraveb 2).
⊘3 Pre	The mixer's aux send (1-4), sub-group (1-4) outputs are accessible from the patch bay.  The mixer's aux return (1-4 stereo) inputs are also accessible from the patch bay.
<b>⊘</b> 4	There is a gain pot for each aux send on the individual channels as well as a master gain for each aux send near the upper right-hand corner of the mixer.
Pan	Aux sends 1 and 2 have a <b>prefade</b> option. If the "pre" switch is down ( on ) for a given channel, that channel's aux send pot operates independently of the channel's fader bar position.
Mute Solo	If, for example, you wanted to process a microphone's signal through the Quadraverb 2 but did not want any of the original unprocessed or "dry" microphone in the mix, you would use the prefade option and route aux sends 1 and 2 to the left and right channels or the Q2. Send the Q2's ouput back to the returns (or two other channels) of the mixer. Leave the mic channels 'fader bar down, adjust the aux send pots for that channel and have the Q2's channel fader bars up.  Aux sends 3 and 4 are always in <b>postfade</b> mode, Both the channel fader bar and the aux send pot must be up to route the signal to the aux send's output. There will always
1-2	be a mix between the dry, unprocessed signal and the processed one.
3-4 Main	To send an individual channel to a sub-group's outputs, check that one of the sub-group switches(1-2 or 3-4) is down (on) for that channel, and that the corresponding sub-group fader bars are up. A channel can be sent to a sub-group out and the main channel out at the same time, or to just one of these outputs.
Default settings	To monitor the main mix out on both the the level meters and headphones, check that the "main mix" swifth (to the left of the headphone pot and meters) is on .You can also monitor sub-cropp out that bere

Delault settings



## Crosspoint Latch Switcher Front Panel Diagram:

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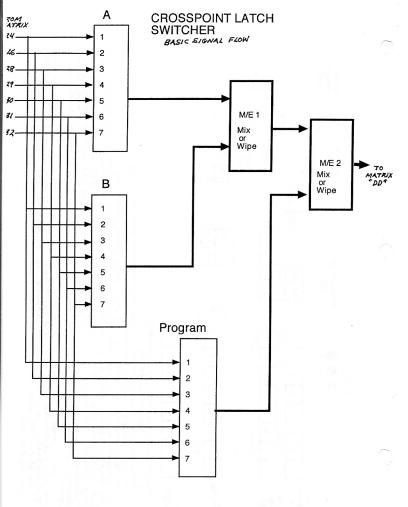
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Channels 1 through 7 are selected at the matrix.

The grey pushbutton can generate either black or a solid field of color depending on the switch above it. When the switch is set to "color", To view individual video signals at the switcher's output, select the signal's channel number on the Program (blue) bus. he knobs next to it adjust the luminance, hue and saturation of the color.

light above "ME1" is on. Select a transition (mix or wipe ) on the ME1 ( top) transition select. If "Mix" is selected the ME1 (top) fader bar will determine the additive mix between bussesA and B. If "Wipe" is selected, the ME1 (top right) wipe pattern select will determine the To mix or wipe between two video signals, select one video signal's channel number on the Mix/Effects A (orange) bus and another channel number on the Mix/Effect B (orange) bus. Then SELECT THE "ME1" PUSHBUTTON ON THE PROGRAM BUS so that the shape between busses A and B and the fader bar will determine the size of the shape or split screen. If the position switch is off, certain shapes (e.g. circles and squares ) will be centered automatically. If the positon switch is on , the joystick will position the shape on the To mix or wice between the composite image from ME1 and a third video signal, select the third signal's channel number on the Program (sibule) bus and select the orange "MEZ" pushbutton or the same bus simularianeous by so that the light above it is on. Then select "Mix" or wither it is on. Then select "there are separate ideal bats, wipe pattern selects, and other controls for ME2. if neither "ME1" nor "ME2" is selected on the Program bus, the switcher will only show an individual video signal (selected from the



### JONES (LUMINANCE) KEYERS 1 AND 2:

### Internal Keying (Two Inputs)

### Monitoring the Individual Signals:

- 1. Select video signals (e.g. TBC's, cameras) for Keyer input channels A and B.
- 2. Send the output of the Keyer to the Output Amp to monitor the signal.
- 3. Set the KEY ON / OFF switch on the Keyer's front panel to the down or OFF position.
- 4. Set the KEY NORMAL / REVERSE switch on the Keyer's front panel to the up or NORMAL position. This setting will show only the video signal being sent to Channel A.
- Adjust Channel A's GAIN and PEDESTAL knobs using both Video Monitor # 1 and the Waveform Monitor as a reference.
- Set the KEY NORMAL / REVERSE switch to the down or REVERSE position. This will show only the video signal being sent to Channel B.
- Adjust Channel B's GAIN and PEDESTAL knobs using both Video Monitor #1 and the Waveform Monitor as a reference.

### Setting the Kev:

- 8. Set the KEY ON / OFF switch to the up or ON position.
- Set the three-position CLIP SELECT switch to either A or B. This determines whether the video input signal for Channel A or Channel B is the clip input or cut-out shape.
- Set the KEY NORMAL / REVERSE switch according to whether you want the dark or light areas of the clip input signal removed to reveal the other signal.

For "Internal " keying ( two images ), there are four possible configurations:

Channel A as the clip input in key normal

Channel A as the clip input in key reverse

Channel B as the clip input in key normal

Channel B as the clip input in key reverse

- 11. Adjust the CLIP LEVEL knob to control how much of Channel A or Channel B's signal you want. The clip level selects the shade of gray of the clip input video signal that will be the "cut-off" point, allowing the second video signal to fill in the gray areas either below or above that cut-off point.
- 12. You can then fine tune the GAIN and PEDESTAL settings for each channel.

### External Keying (Three Input):

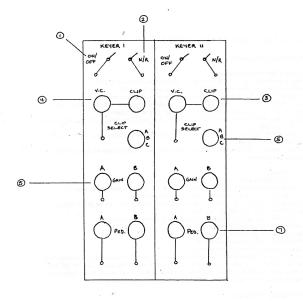
Steps 1 through 8 are the same.

When using a third video signal for the clip input, send that video signal to Channel C of the Keyer by selecting it on the matrix and then set the CLIP SELECT switch for C.

Channel C's input video signal is never seen with all of its gray levels. It functions as a "stencil". By adjusting the CLIP LEVEL knob, this third video signal defines the areas in which either Channel A's or Channel B's video signals are seen, placing them in mutually exclusive parts of the screen. The KEY NORMAL / REVERSE switch will reverse the positions of Channel A and B's video signals relative to the (third) clip input video signal.

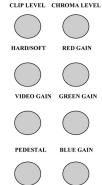
### Using A Color Kill with the Clip Input:

When using a color signal as the clip input, additional colors may result as you adjust the CLIP LEVEL knob. If you want to avoid this, the color kill devices can be patched before the signal reaches the clip input. For example, if you are keying between two color signals, keep those two signals connected to Channels A and B of the Keyer. In addition, send whatever video signal you want to use as a cut-out shape to one of the Color Kills on the matrix. Send the output of that Color Kill to Channel C of the keyer. When you set the CLIP SELECT switch to C, the additional colors will be gone.



- 1. On/Off switch: use or by-pass the clip
- 2. Normal/Reverse switch: type of keying
- 3. Clip level control
- Voltage control input and range adjustment for clip
   Three-position clip select switch
- c. Gain controls for Channels A and B
- 7. Pedestal controls for Channels A and B

Jones Keyer Front Pane



### COLORIZER/MIXER SET-UP

### A ADJUSTING LEVELS FOR THE INCOMING VIDEO SIGNAL

- Set the ON/OFF Switches for each of the six channels to the up or OFF position
- Send a video signal to Colorizer input Channel 1 on the Matrix.
- Send the output of the Colorizer to the Output Amp
- Set the ON/OFF switch for Channel 1 to the down or ON position
- Turn the CLIP, CHROMA LEVEL, RED GAIN, BLUE GAIN AND GREEN GAIN knobs completely to the left.
- Turn the HARD/SOFT CLIP knob completely to the right.
- Turn the VIDEO GAIN knob at around 12 o'clock or straight up.
- Adjust the PEDESTAL knob ,using both Video Monitor # 1 and the Waveform Monitor as a reference. Often the best range for the pedestal is somewhere between 12 o'clock and 4 o'clock, as anything to the left of this range may be all black, and anything to the right may be all white.
- Readjust the VIDEO GAIN knob slightly as needed.

### ADDING COLOR TO THE VIDEO SIGNAL

- Select a color. Turn the CHROMA LEVEL knob straight up at 12 o'clock.
- Turn the RED GAIN knob up. Adjust the CHROMA PHASE knob on the system's Output Amp until the Vectorscope registers the signal in the "R" area. As with all hues in the system, the R,G and B signals of the Colorizer are relative to the Output Amp's CHROMA PHASE settings. Whatever the CHROMA PHASE setting is, these three knobs should represent points 120 degrees apart on the Vectorscope.
- Turn down the RED GAIN knob and adjust the four color knobs (Chroma level for saturation, plus R,G, and B gain controls) for the desired color. The combination of any two of the R, G, and B knobs should give the secondary colors of Cyan, Yellow and Magenta. The example to the right is magenta. Turning up all three knobs may cancel each other out leaving no saturation. The PEDESTAL knob will control the brightness of the color. This color mixes with the input video signal, tinting the entire image that color.

### CLIP LEVEL CHROMA LEVEL HARD/SOFT RED GAIN VIDEOGAIN GREEN GAIN PEDESTAL BLUE GAIN

### CLIP LEVEL CHROMA LEVEL



HARD/SOFT RED GAIN



VIDEO GAIN GREEN GAIN



PEDESTAL BLUE GAIN



### SETTING THE LUMINANCE KEYER

- Set the CLIP SELECT switch for channel 1 to "1".( There are six of these clip select switches at the lower right corner of the Colorizer's front panel. Each of these are six- position switches.)
- Slowly turn Channel 1's CLIP LEVEL knob clockwise. On the Colorizer's oddnumbered channels (1, 3, and 5), the CLIP LEVEL knob will begin taking out the brightest areas of the input video signal, leaving black in its place, and leaving the darkest areas of the video signal colorized. As you turn the knob, the remaining colorized area of the signal may darken, but you can readjust this with the PEDESTAL knob. On the Colorizer's even-numbered channels (2, 4, and 6), the CLIP LEVEL knob will begin taking out the darkest areas of the channel's input video signals, leaving black in its place and leaving the brightest areas of the video signal colorized. As you turn the knob, the remaining colorized areas of the signal may lighten, but you can adjust this with the channel's PEDESTAL knob.

### COLORIZER / MIXER SET-UP continued

### SETTING TWO COLORS FOR ONE VIDEO SIGNAL

- To set up the next channel, turn Channel 1's ON / OFF switch to the OFF position and turn Channel 2's switch ON.
- Send the same video signal to Colorizer input channel 2 (Matrix point # 34.) Repeat the procedure
  for adjusting GAIN and PEDESTAL levels, adding color and setting up the luminance keyer. The only
  difference now is that this keyer will black out the darkest portions of the image, leaving the rest
  colorized.
  - Turn Channel 1 on again.
- Readjust the CLIP LEVEL knobs for each channel as needed. Because the luminance keying circuitry of the odd and even channels alternate the areas of the video signal that are blackened and colorized., adjusting the CLIP LEVELS of the two channels with this patch will allow you to have two separately controllable colors in opposite gray areas of the video image. If both CLIP LEVEL knobs are too far to the left, the two colors from each channel may wash each other out. If the CLIP LEVEL knobs are too far to the right, there may be too much black in the image. After the desired mix between the two channels is established, you may need to readjust the individual PEDESTAL levels again.

### MIXING DIFFERENT VIDEO SIGNALS

• If you chose two different video signals for any two channels of the colorizer, the video signals, and their assigned colors will mix together. The brightest areas of each video signal will come through, with the channels' individual PEDESTAL controls determining the relative strength of each signal. The colors of each channel will combine according to the logic of additive color mixing. For example red from one channel and green from another channel will add up to yellow at the areas of the screen that they are superimposed. Each video signal that you add to each channel will combine at the Colorizer's output, provided that those channels are turned on and the PEDESTAL levels are set accordingly. Often the PEDESTAL knobs for each channel have to be slightly lowered with the introduction of a new channel to the mix. Only leave the ON / OFF switches turned ON for the channels that you are using.

### OTHER VARIATIONS TO THE COLORIZER

- There is MASTER GAIN and PEDESTAL control for its output. These are usually left straight up at 12 o'clock but can be used to adjust the overall luminance values of the Colorizer. Please set them back to 12 o'clock for the next person using the Colorizer.
- There is a separate POLARITY SWITCH for each channel to render the input video signal positive or negative, reversing the gray values. Some pedestal adjustment may be required for each setting. Using these will help in getting more than two colors from a single image source. For example, if you have the same signal going into Channels 1 through 4, you can use the luminance keyers on each channel to separate the grey levels if: Channel 1 is positive ( keys out white ), Channel 2 is positive ( keys out black), Channel 3 is negative ( keys out white) and Channel 4 is negative ( keys out black ). Adjustments of the CLIP LEVEL knobs for each of the channels is important in this patch.
- There are separate CLIP INPUT SELECT switches for each channel. When first using the Colorizer, start with these switches in their default settings (i.e. channel 1 on "1", channel 2 on "2", 3 on "3", etc.) If you do experiment with them, turn them back to their default settings for the next person using the Colorizer. While the luminance keying circuitry in each channel always keys between the (colorized) input video signal and black, these CLIP SELECT SWITCHES allow a separate signal routed from another channel to be used as the cut-out shape, resulting in more complex ways of combining the signals.
- There is also a way to have additive mixing of the channels on the Colorizer. For this you have to set
  the LAYERED/ ADDITIVE SWITCH to ADDITIVE, and have ALL of the channels of the Colorizer
  turned on and ALL of the PEDESTAL knobs turned partially up. Again, if you experiment with this
  one, please set the switch back to LAYERED when you're done. This is the setting used in almost all
  applications of the Colorizer.
- All of the Colorizer's parameters except POLARITY, LAYERED/ ADDITIVE and CLIP INPUT SELECT are voltage-controllable.

### EIGHT CHANNEL SEQUENCER

### Initial Set-up:

- 1. Send video signals to the individual channels (0-7) of the Sequencer through the matrix.
- 2. Send the output of the Sequencer to the Output Amp.
- 3. On the Sequencer's front panel, set the MASTER PEDESTAL knob straight up at 12 o'clock.
- 4. Set the four-position SYNC MODE SWITCH to "1".

### To Monitor and Adjust the Pedestal Levels for Each Channel:

- 1. Set the SEOUENCE / SWITCH control to the down or SWITCH position.
- 2. Set the BINARY / SWITCHER control to the down or SWITCHER position for manual switching,
- In the manual switching mode, the red push-button for each channel will light up that channel's LED and will show the individual video signal at the output amp
- Adjust the individual PEDESTAL knob for that channel using both the Waveform Monitor and Video Monitor # 1 as a reference.

### The Sequence Mode:

- 1. Set the SEQUENCE / SWITCH control to the up or SEQUENCE position.
- 2. Set the COUNT BIAS knob and SYNC BIAS knob straight up at 12 o'clock.
- 3 Send a PULSE wave from the +/ 5 V system to the COUNT INPUT jack of the sequencer. This could be a square wave output from one of the oscillators (but not a sine or triangle output) or a signal with a continuous waveform processed through one of the comparators
  - or a random pulse from the random signal generators or the trigger output of the Korg MS 20 External Signal Processor
- 4. In the SEQUENCE mode, the device will always step through the video signals in the order that they were sent to the matrix starting with channel 0. The rate of the sequence will be determined by the frequency of the incoming +/- 5 v signal (for example, the frequency range switch and fine tuning knob on the oscillator). The last channel stepped in the sequence will be determined by the red push-button for that channel.

### Sync Modes:

- Sync Mode 0: does NOT allow for vertical interval switching ( there will be a visible glitch at the switch point.) It does allow for switching faster than the field rate if the incoming +/- 5 v pulse signal is above 60 cycles.
- Sync Mode 1: **This is the most commonly used mode.** It does allow for vertical interval switching at he frequency of the incoming +/-5 v signal.
- Sync Mode 2: Also allows for vertical interval switching at half of the rate of the incoming +/- 5 v signal.
- Sync Mode 3: Always switches at the field rate of 60 cycles ( or at whatever the frequency of the signal is that is patched into the SYNC INPUT jack. This signal is almost always vertical drive from the Sync Generator ). This mode ignores the signal going to the count input.

### **Switch Mode Logic:**

When the SEQUENCE / SWITCH control is in the up or SEQUENCE position:

- The device is always in the sequence mode and needs a +/ 5 v PULSE signal to the count input
  - The BINARY / SWITCHER control is disabled
- When the SEQUENCE / SWITCH control is in the down or SWITCH position:
   The BINARY / SWITCH control is enabled, allowing a choice between binary control and manual switching.

### **Binary Control:**

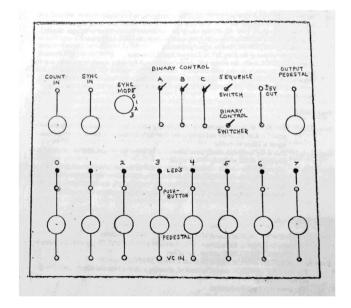
The BINARY CONTROL mode uses the combination of the A, B and C SWITCHES and control voltage inputs to select an individual channel through binary logic

- When all three switches are down, channel 0 is on
- When A and B are down and C is up, channel 1 is on
- When A is down, B is up and C is down, channel 2 is on
- When A is down, and B and C are up, channel 3 is on.
- When A is up, and B and C are down, channel 4 is on
- When A is up, B is down and C is up, channel 5 is on.
- When A and B are up and C is down, channel 6 is on.
- When all three switches are up, Channel 7 is on.

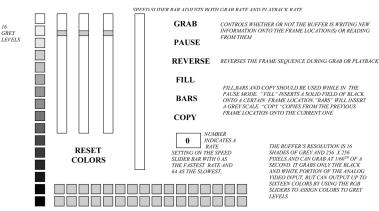
Memorizing binary logic is less important than understanding that when asynchronous control voltages are sent to the A, B and C input jacks, it allows for a more undetermined ordering of images, as opposed to the constant linearity (0,1,2,3...) of the sequence mode.

The A, B, or C inputs can take either a pulse or continuous +/ - 5 v signal

The switch must be up for the corresponding input jack in order for the incoming control voltage to work.



### DESIGNLAB FRAME BUFFER (FB-1) BUFPALLETTE MENU



32 FRAMES OF MEMORY

SPEED

### INITIAL SET-UP

R

- 1. Put the KICKSTART disk into the Amiga 1000 Computer's disk drive.
- 2. When the WORKBENCH prompt appears on the Amiga's monitor, put in the BUFPALLETTE disk.
- 3. Double click on the BUFPALLETTE disk icon.
- Double click on the BUFPALLETTE program icon.
- 5. In the BUFPALLETTE program, select under OPTIONS and FRAMES, "32".
- 6. Send a video signal to the Frame Buffer's (FB-1) video input .
- Send the Frame Buffer's (FB-1) black and white output to the Output Amp.
- 8. On the BUFPALLETTE program menu, click the word GRAB until it is highlighted with a blue rectangle. At this point, the digitized video signal should be seen through the Output Amp.
- 9. Set the SPEED SLIDER BAR to the top setting.
- 10. On the front panel of the Frame Buffer, adjust the BRIGHTNESS and CONTRAST knobs using both the Waveform Monitor and Color Monitor #1 as a reference.

### BASIC FUNCTIONS

There are four basic variations to grabbing ( writing ) and playing back ( reading ) video information with the buffer

- 1. Grab is on Pause is off The buffer will continuously write new information from the incoming digitized video signal to the 32 frame locations at a rate determined by the SPEED SLIDER BAR. The top speed is one sixtieth of a second. As it loops through the 32 frames, it will continuously erase the old information that was in the frame location and update it with a new image.
- 2. Grab is off Pause is off The buffer will read or play back the last 32 frames in memory at a rate determined by the SPEED SLIDER BAR.

  3. Grab is off Pause is of The buffer will read one frame at a time and hold it at the output until another frame location icon is clicked.
- B. Grab is off Pause is on The buffer will read one frame at a time and hold it at the output until another frame location icon is clicked. You can randomly access any of the 32 frames
- 4. Grab is on Pause is on The buffer will grab or write one frame at a time. To do this:

  Click on one of the 32 frame location icons where you want to hold the image. Put the speed slider bar in the top position. As soon as you see the image that you want, one of two methods can be used:

- Deselect GRAB. You can see the single frame that you've just grabbed Go to the next frame location that you want to write on Select GRAB and repeat this process
  - Staying in GRAB, click to the next frame location. You won't see the still image that you've just grabbed until later.

With either method, when you're done selecting and grabbing the frame sequence that you want, **Deselect GRAB and then deselect PAUSE** to play back the sequence.

If you deselect PAUSE first, you will inadvertently erase your sequence. Adjust the SPEED SLIDER BAR for the playback rate.

### SELECTIVE STORAGE WITH THE BUFFER'S LUMINANCE KEYER

The buffer has a built-in luminance keyer and a separate video clip input on the matrix, but rather than keying between two different incoming video signals as in a conventional keyer, the buffer is keying between different data. That data is either in the form of old information that has already been written into the buffer's memory or new information that it is currently writing into memory.

When the Buffer's KEY ON / OFF SWITCH is down or in the OFF position, the keyer is disabled. The buffer is then either grabbing( writing) a full frame or playing back (reading) a full frame of video information at any given time. With the KEY ON / OFF SWITCH in the up position and a video signal sent to the FB-1 clip input, it is capable of writing, or updating a selected portion, or so many pixels of a frame location. This selection is determined by: 1. the grey levels of the clip input's video signal, 2. whether the front panel's KEY NORMAL / REVERSE switch is up or down, and 3. the setting on the CLIP LEVEL knob. In all cases, the buffer is always writing information from the main input's video signal, but only onto specific areas of the frame location designated by the clip input's video signal. There can be several variables to this process. Among them:

Grab On	Pause On	Key Normal	The buffer will continuously update video information onto areas of a single frame location designated by the darkest portions of
			the clip input's video signal.
Grab On	Pause On	Kev Reverse	The buffer will continuously update video information onto areas

- of a single frame location designated by the lightest portions of the clip input's video signal.
- Grab On Pause Off Key Normal The buffer will continuously update video information onto areas of each successive frame location designated by the darkest portions of the clip input's video signal at a rate determined by
- Grab On Pause Off Key Reverse the "Speed" slider bar.

  The buffer will continuously update video information onto areas of each successive frame location designated by the darkest portions of the clip input's video signal at a rate determined by

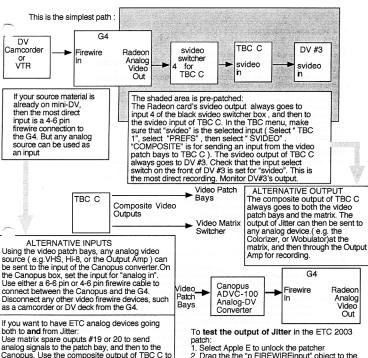
the "Speed" slider bar.

Another important variable is whether or not there are two different video signals going into the video and clip inputs or the same video signal.

### COLOR MAPPING:

There are two outputs to the buffer. "FB-1 Black and White" will always be monochromatic and have a somewhat sharper signal. "FB-1 Color Map " will show different hues and saturation values for each grey level as they are assigned with the R G B slider bars in the Bufpallette program menu.

16- color palettes can be composed in Deluxe Paint using the color spread, copy and exchange functions and imported to the buffer. In Deluxe Paint, save the file under "Save Picture" to an Amiga-formatted floppy disk and bring it into Bufpallette under the "Load Picture" selection. You can then grab new images with this palette.



If you are using the ETC 2003 patch, test the firewire input:

Output Amp as the final record out.

send Jitter back to devices at the matrix. Use the

- Double-click on the "p FIREWIREinput" object
   Click "getdevlist". "DVvideo" should appear under
- "prepend append".

  3. Click "open"
- lick the toggle above "metro 2" object, so that a. . . appears.
- The live firewire image should appear in the small window.
- Once you have this, you can test the output of Jitter

- Drag the the "p FIREWIREinput" object to the white work area.
- 3. Connect the (bottom left) outlet of the object to the (top left) inlet of the "jit matrix 4 char 320 240" object
- object. 4.Press Esc on the keyboard . The image should be
- full screen on the computer monitor.

  5. Press ESC again. The monitor will return to the ETC patch, with a small floating window of the
- image.

  6. Drag this window to the right of the desktop and off the screen until it appears at the output of TBC.
- 7. Select ESC again. This is the video output for Jitter.

### +/- 5 VOLT ANALOG SYSTEM

### APPLICATIONS:

- 1, as control voltages to any of the following video devices:
- Colorizer/Mixer ( Gain, Pedestal, Clip Level, Chroma Level, Red. Green, Blue Mix)
- Paik-Abe Raster Scan ( H.V and S inputs )
- Luminance Keyers (Gain, Pedestal, Clip Level, Key On/Off, Key Normal/Reverse)
- Designlab Frame Buffer (Gain, Pedestal and Clip Level) Mixer (Gain/Polarity)
  - Eight-Channel Sequencer ( Count, Individual Channel Pedestals, Master Pedestal )
- Output Amplifier: Gain and Pedestal 2. as audio signals via the [+/-5V to audio line converters], which send the signals to the
- audio patch bay 3. as video signals via the [+/- 5V to video line converters] which send the signals to the
- video matrix
- 4. as signal sources and control voltages to other +/- 5 volt modules
- 5. as MIDI data via the J.L. Cooper Fader Master

### TYPES OF MODULES IN THE +/- 5V SYSTEM:

Voltage Controllable Oscillators, Positive and Negative Signal Generators:

Ramp Wave Generators, Triangle Wave Generators, Random Signal Generators ( white noise, random pulse,

random ramp etc.)

Signal Processors:

Changing gain: Attenuators, Dual Voltage - Controllable Amplifier

Changing polarity: Inverters, Inverter X 2

Changing waveshape: Comparators, Voltage-Controllable Filter, Waveshaper

Combining signals: Mixers, Sample and Holds, Dual VCA, Splitting signals: Multiples

### PARAMETERS OF THE SIGNAL

A waveform is a graphic representation of an electronic signal with time represented in the horizontal axis

and voltage in the vertical axis Gain: the total voltage excursion or sweep from high to low. In audio this is often a function of amplitude and in video it is a function of grey levels. As a control voltage to a device, it

determines how much of an effect the signal will have on the parameter it is applied to. Gain can be increased with an amplifier or reduced with an attenuator Frequency: the number of times that a waveform completes one cycle from low voltage to high voltage and back in a finite period of time. The unit of time is usually one second, and "cycles per second" are expressed as "hertz". Control voltages are often referred to as "slow-varying", a general range from one cycle every couple of minutes to frequencies directly below the audio spectrum. Audible frequencies can be from 16 to 20,000 Hz. for humans with higher frequencies representing higher pitches. In video, the field rate (60 Hz.) is an important threshold frequency for it is at this point that changes in voltage are represented graphically by changes in gradations along the vertical axis, with frequencies at multiples of the field rate producing more horizontal bands on the raster. The next important threshold is the line rate or 15,750 Hz (a number derived from the multiplying the number of lines in a frame, 525, by he number of frames in a second, 30). Above this frequency changes in voltage, or brightness

take place within each scan line of video. Waveshape the nature of the change in voltage with respect to time. There are two broad categories of waveshapes,: continuous (including sine and triangle), which have gradual transitions between low and high voltages; and pulse (including square) which jump instantly between high and low voltage, although some waveshapes, such as positive and negative ramp ( or sawtooth ) waves have characteristics of both types. As a control voltage, the waveshape will determine whether the change to the applied parameter will be gradual or immediate . As an audio signal, it will be represented by the timbre of the sound.

Signal	APPLICAT	IONS OF + / 5 VOLT SIGN	ALS
Generators	As Slow-Varying Control Voltages	As Audio Signals	As Video Signals
FREQUENCY	the rate of change of the applied parameter often <60 HZ for control voltages to video devices often <20 HZ for control voltages to audio devices	Pitch Audio Frequency Range: ~20 Hz- 20 Khz an octave above a given tone is twice the frequency	>60 Hz ( Field Rate ) for generating horizontal shapes >15,750 Hz ( Line Rate or 525 lines X 30 frames/sec ) for generating vertical shapes
ROUTING	patch directly to the control voltage input of devices, using mini cables	patch to the +/-5v to audio converters ( sends the signal to the audio patch bays, where it can be routed to the mixer and/ or any audio processor)	patch to the +/-5v to video converters ( sends the signal to the matrix switcher where it can be routed to any video device )
GAIN	the overall sweep or range of effect on the applied video/ audio parameter; controlled by attenuators	the overall volume of the audio signal	the overall contrast of the video signal high voltage is white. low voltage is black
WAVESHAPE  Continuous (Sine, Triangle)	the nature of change on the applied parameter with respect to time sine waves as control voltages result in the most gradual transitions in a parameter change, used for dissolves or fades	timbre, tone color Sine waves contain only the fundamental frequency, with no overtones, and form the building blocks for more complex sounds.	determines how the range of gray levels is spread across the norizontal or vertical axis a sine wave just above 60Hz will form a continuous range of gray levels from the top of the video raster to the bottom
Pulse	the most abrupt changes in a parameter, alternating between 2 discrete states, analogous to a switch	already a complex waveform with several harmonics, resulting in a richer tone.	a pulse wave just above 60 Hz will produce only two gray levels from the screen's top to bottom., resulting in hard-edged white and black horizontal bars
OTHER CONSIDERATIONS	Bias is the center point of the sweep, often determined by the potentiometer for the parameter being controlled (e.g. for a CV of a fixed gain, use the pedestal knob of the colorizer to control whether it sweeps from black to full video, or full video to white.  Vertical Interval Switching A Sample & Hold patch may be used to clean up glitches that sometimes occur when using a CV on video devices.  Synchronizing Controls with a control when using a CV on video devices and be control voltages can be	Additive Synthesis Simple waveforms are combined for more complex sounds through mixers & the Dual VCA; also by frequency modulating oscillators with control voltages that are in the audio frequency range Subtractive Synthesis Complex waveforms are broken down usually through VC Filters by removing partials from their frequency spectrum Envelope Generators An envelope is a contour ( shape) that is used to create some of the transient (changing) characteristics of the sound; usually express- ed as Attack, Decay, Sus-	Sync Inputs The sync inputs on the oscillators are most likely to be used here. For signals >60Hz but-15.75 Khz, vertical drive is often patched to the sync input to synchronize the beginning of a field with one of the cycles (i. e. to make the bars stop rolling). For signals above the line rate, horizontal sync is most likely to be used to synchronize the beginning of each line with one of the cycles of the waveform. But free-floating oscillators and random signal generators are also used. Complexity 2 or more oscillators ( usually one synced to vertical and one to horizontal ) are often mixed together to form more complex

### + / - 5 Volt System: Continuous and Pulse Waves

### Continuous

Continuous waves include most waveforms that are not a pulse (or random) wave. They can be symmetrical ( sine or triangle waves) or asymmetrical (positive and negative ramp waves, which are only continuous on one side.) Sine waves are the most continuous and simplest type of waveform, containing only the fundamental frequency, offering the smoothest transitions when used as a control voltage. Ramp waves allow either a gradual rise and sudden fall in voltage (positive) or sudden rise and gradual fall (negative). They can also be contours determined by other means (e.g. Envelope Followers. determined by fluctuating audio levels and Envelope Generators, determined by the parameters Attack, Decay, Sustain and Release.)

Sources of Continuous Waves triangle and sine wave outs of Oscillators

Korg MS20 ESP: audio to envelope out Ramp Wave Generators signals through VC Filter and Waveshaper Doepfer Envelope Follower Envelope Generator of the Korg MS20 Doepfer Light to Voltage Source Doepfer MIDI to CV out

Applications: parameters to modulate gain of OA, Colorizer, Keyer, FB-1 pedestal of OA, Colorizer, Keyer, FB-1, Sequencer key clip level of Keyer, Colorizer, FB-1 red green, & blue gain, and chroma level of Colorizer H, S & V control of Wobulator frequency modulation of Oscillators control of gain on VC Amplifier or VC Mixer control of cut-off frequency on VC Filter inputs to JL Cooper Fadermaster for MIDI

Ways of Processing/ Coordinating Signals
Multiples to spilt signal
Sample & Hold for stair-stepped voltage
Comparator to change to a pulse wave
Inverter to change polarity
Morph Controller for multiple delayed outputs
Mixing through basic or VC Mixers

### Pulse

A **pulse wave** is any waveform that is either high voltage or low voltage with no stages in between.

The pulse width is the ratio of time that a pulse signal is high voltage to low voltage, sometimes referred to as its duty cycle. A comparator can be used for pulse width modulation (PWM)
A square wave is a type of pulse wave that is symmetrical, or high voltage and low voltage, each 50% of the time.

volfage, each 50% of the time. The terms trigger, gate or clock are often used as applications of a pulse wave (for example, Envelope Generator trigger inputs, Envelope Follower trigger outputs, Sample and Hold clock inputs, Clock Divider etc.)

Sources of Pulse Waves pulse out of the Ramp Wave Generators

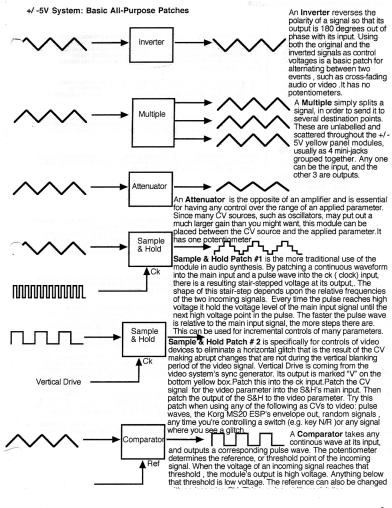
Random Pulse Generator square wave outputs of Oscillators Korg MS20 Ext Signal Proc: audio to trigger out signals through Comparators clock out of the Doepfer MID to CV

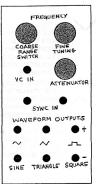
### Applications

Key nor/ off Key normal/ reverse Sequencer count in Sequencer binary controls in clock input to a Sample& Hold step in to Random Sample & Hold trigger input to Envelope Generators trigger input to Ramp Wave Generators

Ways of Processing/ Coordinating Signals Multiples to split signal

Clock Divider for multiple rhythmic pulse outputs Morph Controller for multiple delayed outputs Inverter to change polarity





THE FREQUENCY OF THE WAVEFORM IS CONTROLLED BY A COMBINATION OF THE COARSE RANGE SWITCH AND FIRE TUNNED KNOB. THESE ALLOW FOR A COMPLETE RANGE OF APPLICATIONS, FROM VERY SLOW VARYING CONTROL VOLTAGE S OF LESS THAN A CYCLE PER MINUTE, THROUGH THE AUDIO SPECTRUM TO MULTIPLES OF THE VIDEO LINE RATE FOR PRODUCING SHAPES ON THE VIDEO RASTER.

THE FREQUENCY OF THE WAVEFORM CAN BE MODULATED BY AN INCOMING CONTROL VOLTAGE FROM OTHER MODULES IN THE +/-5 V SYSTEM. THE ATTENUATOR KNOB DETERMINES HOW MUCH OF AN EFFECT ON THE FREQUENCY THE INCOMING CONTROL VOLTAGE WILL HAVE.

THE SYNC INPUT ALLOWS FOR SYNCHRONIZING THE BEGINNING OF EACH CYCLE (
OR RVERY SO MANY CYCLES) OF A WAVEFORM WITH OTHER TIMING STRUCTURES IN
THE SYSTEM SUCH AS VERTICAL DRIVE, HORIZONTAL DRIVE, OR A PULSE WAVE
FROM ANOTHER MODULE

THERE ARE SIX OUTPUTS ON THE OSCILLATOR; THREE WAVESHAPES (SINE, TRIANGLE AND SQUARE), EACH WAVESHAPE HAS A POSITIVE AND NEGATIVE OUTPUT, WHICH ARE ALWAYS 180 DEGREES OUT OF PHASE WITH ONE ANOTHER.

### SAMPLE AND HOLDS



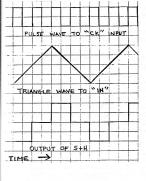
SAMPLE AND HOLDS COMPARE TWO SIGNALS, ONE AT THE CLOCK INPUT AND ONE AT THE MAIN INPUT AND PRODUCE A CORRESPONDING OUTPUT. WITEN THE #-5 V SIGNAL BEING SIBNT TO THE "CK" INPUT IS AT HIGH VOLTAGE, IT WILL SAMPLE THE VOLTAGE LEVEL OF THE MAIN INPUT'S SIGNAL AND HOLD IT AT A STEADY LEVEL UNTIL THE INEXT HIGH VOLTAGE COMES INTO THE "CK" INPUT, WITEN IT WILL SAMPLE AND HOLD AGAIN. THERE ARE AT LEAST TWO APPLICATIONS OF THIS:

- A STAIR-STEPPED CONTROL VOLTAGE WILL RESULT WHEN A CONTINUOUS (E.G. TRIANGLE OR SINE) WAVE IS SENT TO THE MAIN "IN" AND A PULSE ( E.G. A SOUARE) WAVE IS SENT TO THE "CK".
- 2. VERTICAL INTERVAL SWITCHING WITH A CONTROL VOLTAGE IF A PULSE WAYE IS SENT TO THE CONTROL VOLTAGE INPUT OF A VIDEO DEVICE (SUCH AS A SQUARE WAVE TO THE COLORIZER'S PEDISTAL LEVEL.) OR ANY WAVESHAPE IS SENT TO A CONTROL SWITCH (SUCH AS THE KEY NORMALÆVERSE CONTROL ON KEYER I), THERE WILL BE A VISIBLE "GILICIET" IN THE VIDEO SIGNAL WHEN THE CONTROL VOLTAGE CHANGES. THIS IS BECAUSE THE CHANGE IS NOT TAKING PLACE DURING THE VERTICAL BLANKING PERIOD.

SENDING THE CONTROL VOLTAGE. TO THE MAIN "IN" OF THE SAMPLE AND HOLD FRST WHILE SENDING VERTICAL DRIVE (FROM YELLOW BOX.# 2) TO THE "CK" INPUT WILL RESULT IN AN OUTPUT THAT SWITCHES CLEANLY, OR IS SYNCHRONIZED WITH THE VERTICAL BLANKING PERIOD OF THE VIDEO SYSTEM.

THE SAMPLE AND HOLD MODILES ON YELLOW BOX #1 OFTEN WORK IN CONJUNCTION WITH THE TRIANGLE WAVE GENERATORS TO THEIR RIGHT AND IS CONFIGURED SOMEWHAT DIFFERENTLY THAN THE ABOVE MODULE. THE PULSE IS SENT TO THE "S" INVIT AND ITS OUTPUT IS SENT TO THE "H" INVIT OF THE TRIANGLE WAVE GENERATOR.

THERE IS ALSO A RANDOM SAMPLE AND HOLD GENERATOR ON THE SAME BOX. WHEN A PULSE IS SENT TO IT, THE "STEP" OUTPUTS SEND A RANDOM VOLTAGE LEVEL WITH EACH PULSE SAMPLE.



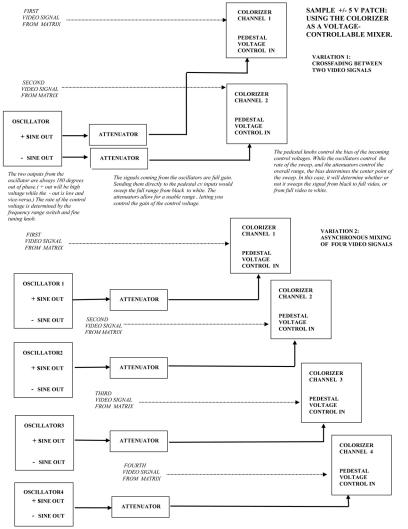
NEGATIVE RAMP WAVE GENERATOR		ATTENUATORS & MULTIPLES WAVESHA	SIGNAL GENERATORS PER SAMPLE & H	TRIANGLE WAVE OLDS GENERATORS
MP WAVE GEN	+ RAMP WAVE GENERATOR	ATTENUATORS OF GAIN	WHITE MOISE	TRIANGLE WAVE GEN.
PULSE RAMP OUT	VC ATTEN SUSTAIN PLOUT PA OUT	IN IN SHAPE	1	PULSE OUT TRI OUT
YC PULSE LED	VC YC SINE OF	OUT OUT VC	PAMP OUT	VC HOLDIN LED
MULTIPLE	INVERTER XA	MULTIPLE VC	ONE	PULSE MANUAL GATE
-RAMP WAVE GEN	+ RAMP WAVE GENERATOR	• • •	STEP OUT OUT S	TRIANGLE WAVE GEN
• •		MULTIPLE OUT	TREMELO PULSE	• •
• • 0	• • • •	• • • out	OUT OUT OUT	

POSSITIVE

RANDOM

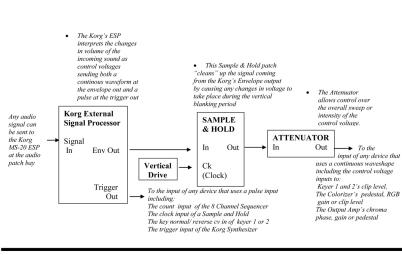
co	VOLTAG NTROLL SCILLAT	ABLE		CAPACITOR, VERT ICAL & HOR IZ. DRIVE, INVERTER, COMPARATORS	ATTENUATORS & MULTIPLES	MIXERS & MULTIPLES	VOLTAGE- CONTROLLABLE ( FILTER	DUAL VOLTAGE- CONTROLLABLE AMPLIFIER
VCO FREQ	Ľ,	O VCO	r	CAPACITOR	ATTENUATORS	MIXER IN	HI CO	vc PAN
FINE	H.SYNC V,SYNC		r	VERT D. HORIZ D.	N IN	• 2 • outs	Q PRES	O O
VG ATTEN	EXT SYNC		•	IN OUT	TUO TUO	● 3 ● QUTS ●	VC VC ATTEN, ATTEN,	1 IN 2 IN
Ac .	LIN	•	•	REF IN	MULTIPLE	MULTIPLE	VC VC	1+2 1+2 IN VC
•ıv/	ост	•	•	COMPARATOR	• •	• 1	LIMIT MAIN	1 VC 2 VC
RAI	•	•	•	REF IN	MULTIPLE	● 3 ● OUTS ●	HP BP	1 2
TRI	SINE	1000			J	10.000 10.000 10.000		1170 17

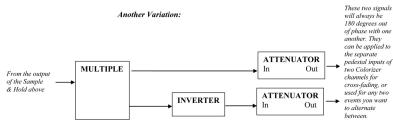
+ / - 5 VOLT ANALOG SYSTEM MODULES



### SAMPLE PATCHES FOR TURNING AN AUDIO SIGNAL INTO A CONTROL VOLTAGE

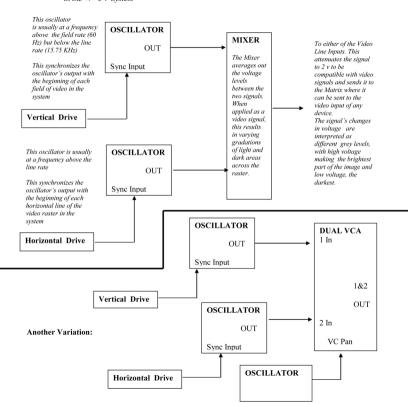
 Vertical Drive, Sample & Holds, Attenuators, Multiples, and Inverters are all modules in the +/-5 V Analog System





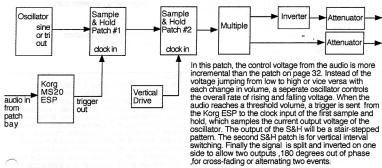
### SAMPLE PATCHES FOR TURNING +/- 5 V SIGNALS INTO VIDEO SIGNALS

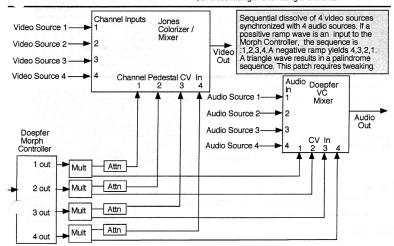
Oscillators, Mixers, Vertical and Horizontal Drive, and the Dual VCA are all modules in the +/- 5 V System



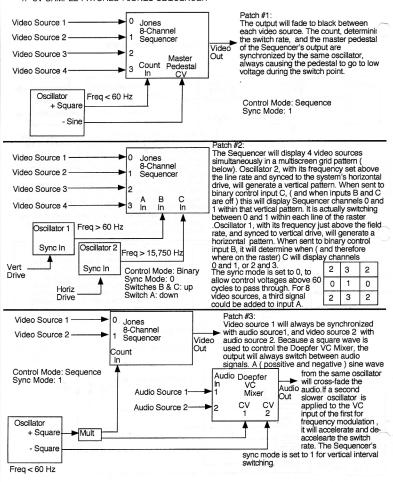
### + / - 5 V SAMPLE PATCHES: More Audio to CV / Colorizer & Morph Control

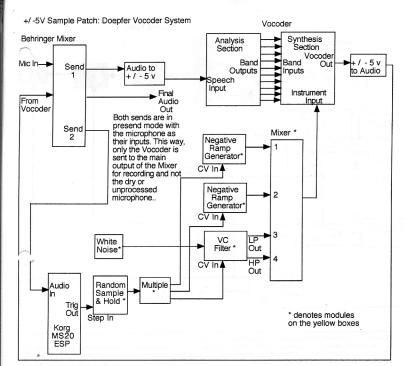
### Audio to Control Voltage Alternate Patch





### +/ -5V SAMPLE PATCHES : JONES SEQUENCER





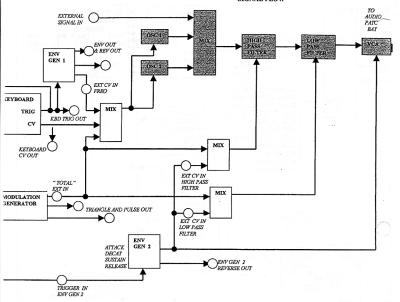
There are many variations to ( and a seperate manual for ) the Doepfer Vocoder System. This patch only uses the two main modules, the Analysis and Synthesis sections, with each of the band outputs going to their corresponding inputs ( Low Pass to Low Pass, Band 1 to 1 etc.). In this case, the Vocoder is treated as a black box, with two inputs ( speech and instrument) and one output.

The microphone is sent to the speech input, first through the Behringer mixer, and then out of Send 1 to the audio +/-5v converter. This adjusts the signal to the best levels for the Vocoder Use the send's gain pots on the mixer to fine tune the incoming levels to the Vocoder. The best adjustment is when all of the led's in the Analysis section light up when speaking, and shut off when there is no sound.

The best results are when there are fairly complex waveforms sent to the instrument section. A single sine or triangle wave, for example, would not be effective. This patch uses three sources, two ramp wave generators, with frequencies set in the audio range, and filtered white noise. The relative levels of the three hals are adjusted through a seperate mixer.

ne same microphone source is being sent to the Korg' MS20 ESP, which triggers a random S&H module. This is the main control for the three sources above, frequency modulating the two ramp waves, while changing the cut-off frequency of the VC Filter.In this way, every time there is a change in volume from the voice, there is a change in the parameters modulating the voice.

# KORG MS-20 AUDIO SYNTHESIZER SIGNAL FLOW



Unlike the + / - 5 V system, the Korg Audio Synthesizer has a default patch :

Oscillators I and 2 are the primary sources for audio signals in the Korg. There are front panel controls for the waveshapes, octaves and tuning of each oscillator. The relative levels of the oscillator signals are controlled at the mixer where an external audio signal can be added from the audio patch bay or the +/ - 5 V system. The High Pass Filter will allow only the high frequencies of the signal through. A cut-off knob determines the threshold frequency. There is also a Low Pass Filter for removing all but the lowest frequencies. Finally a CVCA, or Voltage-Controllable Amplifier determines how dud the signal will sound and the Envelope Generator (# 2) controlling it determines how this loudness changes in respect to time, with separate controls for parameters Attack, Decay, Sustain and Release. Pressing any key on the keyboard will trigger Envelope Generator 2. The keys will also modulate the frequency of Oscillators I and 2. Envelope Generator 2 can also be used to control the cut-off points of the High and Low Pass Filters. The display of the Company of the High Pass and Low Pass Filter cut-offs, and to frequency-modulate Oscillators I and 2.

The Korg allows for external control voltages to be applied at several points in this patch to either add to or bypass pre-existing control voltage sources in the Korg. For example, patching a pulse wave from the +1-5 V system to the TRIGGER IN] pack will replace the keyboard as the trigger source to Envelope Generator 2. Patching an oscillator from the +1-5 V system to the TOTAL EXT IN jack will replace the Modulation Generator as a control for the Filters and Oscillators 1 and 2. The KEYBOARD CV OUT jack can be patched to a device in the +1-5 V system to modulate its parameters with different notes on the keyboard. Several modules exist independently on the MS-20, including a SAMPLE AND HOLD, ENVELOPE GENERATOR I, WHITE NOISE GENERATOR and the EXTERNAL SIGNAL PROCESSOR that can interface with both the Korg's main synthesizer and the +1-5 V system.



THE Q2 has left and right inputs and outputs on the audio patch bays. You can patch a sound source directly this way, but usually the Behringer Mixer's sends, or auxiliary outputs, are sent to the inputs of the Q2. This way, any combination of audio signals being sent to the mixer can be routed to the Q2. The Q2's outputs are then sent to one of the mixer's main inputs, or its returns. There are several places to adjust levels for each (left and right) channel: the individual sends for each of the 16 mixer channels, the master send levels, the Q2's inputs levels and the Q2's output levels.

The Value / Enter control is both a rotary dial and a push-button, and is the main interface for the Q2.

When Program is selected, Value / Enter scrolls through the 100 factory presets and 100 user settings.

A Program is a stored patch represented by a flow chart in the display

A **Block** is the equivelant on an analog sound module. Up to 8 Blocks can be used in a program, memory permitting. ( When the Q2 is out of memory, the display will read "DSP is full").

There are four basic categories, or Types of blocks: Pitch, Equalization, Delay and Reverb.

Each of these types have several sub-types.

A Pitch block can be a Flanger, Phasor, Lezlie ,Pitch Shifter , Pitch Detuner ,or Ring Modulator An Equalizer block can be a Low ,High or Band Pass Filter, Resonator, Tremolo, Phase Inverter, Graphic or Parametric Equalizer

A Delay block can be a Mono, Stereo or Ping-Pong Delay or a Sampler.

To change a block type:

In the display is a small black triangle above one of the blocks. Use the left or right arrow of the *Block* select to scroll to the block you want.

"lect Type. The heading in the display wil show the block number and the type.

the \*Value/ Enter control to scroll through the 5 selections: Off, Equalization, Pitch, Delay and Reverb. When you find the desired type, push Value /Enter to select it. The sub-type selected next. For example, If you selected Pitch the heading will now read "pitch type", and you can scroll through the options (mono chorus, stereo chorus etc.), selecting one by pushing Value/ Enter.

Each selection has several parameters, depending on its type.

Select Parameter. The current parameter to be edited will have a numerical value that is underlined and to the right of the parameter name. Use the Value/Enter dial to set a numerical value and push Value/Enter to select it. The next parameter will now be underlined and ready for editing.

The Routing select will allow you to scroll through various routing possibilities between the inputs, outputs and blocks, each shown in the display's patch diagram. The Mix select will allow the control of levels to each

block.

in any of the Block modes ( Type, Routing, Parameter and Mix ), you can use the left and right Block arrows to move to the block that you want to edit.

Toggling Compare on and off allows you to hear the original patch before editing and compare it to what you

currently have set.

When Store is selected, you can scroll to the user preset number that you want. Pushing Value / Enter will save the current settings to that location. It will not erase over factory presets.

MIDI is used to select a MIDI channel (leave on 1), MIDI thru (leave off) etc.

Modulation allows you to determine what MIDI parameter will control a specific Q2 parameter in real time and by how much.

Select Modulation.

With Value/Enter, select a modulator (MOD 1-8)

Select the block that you want to control.

Select the specific parameter of that block to be controlled.

Select the modulation source (1-8).

act the MIDI controller. This is the MIDI parameter( Pitch Bend , Aftertouch, Note Number, Continuous Controller select a numerical value).

Make sure that the ETC MIDI router has a device selected that is transmitting MIDI data to the Q2 (e.g. the JL Cooper Fadermaster, G4 w/a MAX patch, or the keyboard), and that this device (or software) is set for the same MIDI channel number and parameter that you've selected for the Q2.

SAMP	73 SAMPLE TIME ADJ.	74 IMPUT FILTER FREQ.	75 LINE/MIC LEVEL INP
SAMPLING	The WAVESAMPLE parameters apply	sample. The SAMPLING CONFIG.	FIGURATION parameters can be

г	SAMPLING CONFIG;	JG CONFIG;
yldd:	73 SAMPLE TIME ADJ.	76 SAMPLING THRESH
-16.	74 INPUT FILTER FREQ.	77 USER MULTISAMPI
_	75 LINE/MIC LEVEL INPUT	2016 (301 (301 (311)

76 SAMPLING THRESHOLD	1.1 SAVE LOWER KBD. SOUND 14 SAVE CONFIG. PARAMS	14 SAVE CONFIG. PARAMS
77 USER MULTISAMPLING	12 SAVE UPPER KBD. SOUND	15 COPY PROGRAM TO LOWER
2 (2) (3) (3)	13 SAVE BOTH KBD. SOUNDS	16 сору ряосяли то иррея
		91

COMMAND

CONFIGU	CONFIGURATION	
31 MIDI OMNI MODE	86 EXT. CLOCK JACK SELECT	26 WAVESAMPLE SELECT
32 MIDI CHANNEL SELECT	87 INTERNAL CLOCK RATE	27 INITIAL WAVESAMPLE 01-08
33 MIDI THRU MODE	88 SEQUENCER LOOP SWITCH	60 WAVESAMPLE START
34 MIDI CONTROLLER ENABLE   89 SEQ. FT. SW./SUS. PEDAL	89 SEQ. FT. SW./SUS. PEDAL	61 WAVESAMPLE END
35 EXT. SEQUENCER CLOCK		62 LOOP START 00-FF

26 WAVESAMPLE SELECT	63 LOOP END	68 RELATIVE TUNING—FINE
27 INITIAL WAVESAMPLE 01-08	<b>64</b> 100P END FINE ADJ. 00-FF	69 RELATIVE AMPLITUDE 00-63
60 WAVESAMPLE START	65 toop switch	70 RELATIVE FILTER FREQ.
61 WAVESAMPLE END	66 WAVESAMPLE ROTATE	71 MAXIMUM FILTER FREQ.
62 LOOP START 00-FF	67 ВЕГАТІЧЕ ТИМІМВ—СОАВІЗЕ	72 TOP KEY 01-61.

	29	~	3
PROGRAM	The values of these program	BOARD) can be saved on diskette	as any of four programs assignable to each sound.

36 FILTER CUTOFF FRE 00-99	37 FILTER RESONANCE 00-40	32 LED DEPTH OG. Wheel 38 FILTER KBD. TRACK 00-99, 00-Mod. Wheel 38	
29 MONOPHONIC MODE	31 LFO FRED. (SPEED)	LFO DEPTH 00-99, 00=A	
23	3	32	

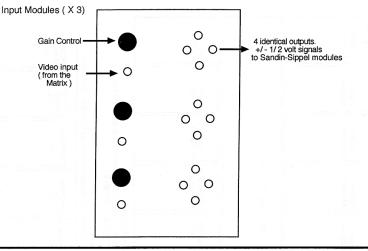
KEYBOARD/PROGRAM

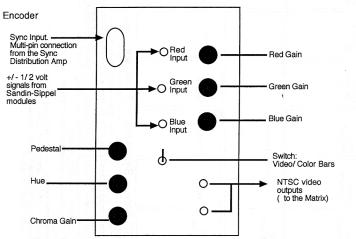
NGE SENS.		C-1 00-99	C4 00-63
93 KEYBOARD VEL. SENS.	22	PITCH BEND RANGE 00-34	25 U/L PROGRAM LINK
E9-00-93	23	KEYBOARD VEL. SENS. 00-63	

= GENERAL KEYBOARD ===

ENVELOPE CONTRACTOR CONTRACTOR MODULATION	1	770CX 00-31 50 ATTACK 45 ATTACK—VEL SENS. 55 ATTACK—VEL SENS. 55 ATTACK—VEL SENS. 00-31	51 PEAK	52 DECAY. 31	53 SUSTAIN 48 SUSTAIN—VEL. SENS.	erit erit erit
ENVELC	FILTER	40 ATTACK 500-31 50	130	42 DECAY 52		or it
WAVESAMPLE	20 01-08	27 INITIAL WAVESAMPLE	28 MIX MODE	33 оѕс. 2 ветине	34 05c. MIX	35 0SC. MIX—VEL. SENS.

			_	
Adder/ Multiplier		Adder/ Multiplier	ledd	Reference Modules (9)
Adder/ Multiplier	,	Adder/ Multiplier	d LFO Bank	Comparator
Adder/ Multiplier	ars ( 5 )	Adder/ Multiplier	be Follower an	Oscillators (2)
Adder/ Multiplier	Mini Adder/ Multipliers ( 5)	Ampitude Classifier	out Sippel Envelop 1 through 6 ) n Generators,	Oscillators (2)
Adder/ Multiplier	Mini /	Ampitude Classifier	Sandin IP Lay (ey Stacker; S Matrix outputs in IP. ir output.	Differentiator (3)
Ampitude Classifier		Comparator	Sank Sippel K Sank Sippel K of the Sandin (I s on the Sand signals at the titors, Differenti	to are the uputs and Vatrix
Function Generator (3)	4)	Encoder	ETC Sandin-Sippel Image Processor Main Sandin IP Layout Not shown: Sippel Mix Master: (4) 9-Input Mixer Bank; Sippel Key Stacker; Sippel Envelope Follower and LFO Bank There are six outputs from the ETC Matrix switcher to the Sandin (Matrix outputs 1 through 6) These can be patched to any of the six Input Modules on the Sandin IP. These modules stip sync and produce +1-1/2 V signals at their output. All of the other modules (Adde-Mutipliers, Comparators, Differentiators, Function Generators, Amplitude Classifiers and Sippel	patchable in many configurations.  The final modules in the system that signals are sent to are the encoders. Each of the encoders take a+-1/2 V signals at their inputs and produce an NTSC video signal at their outputs.  The outputs can be sent to three inputs on the ETC Matrix (inputs BB,CC and DD)
Input Modules (3)	Mini Adder/ Multipliers ( 4 )	Encoder	pel Image Master: (4) m the ETC Ma o any of the sin c and produc c and produc sin the fill	gurations. system that si ce +/- 1/2 V s o signal at thei t to three input D)
input Modules (3)		Encoder	andin-Sip n: Sippel Mix six outputs front be patched to dules strip syr dules strip syr process signes	in orders against the incoming and incoming
Sync Generator	Sync Distribution	Stripper	ETC Solution Not show There are These car These car These mc All of the Columbia Col	patchable The final mencoders. Each of the produce a The output (inputs E





# Adder / Multiplier

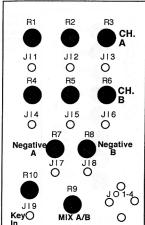
The Adder Multiplier is used to add ( superimpose), f and gain control (

i...dply) signals.
JI1, JI2, JI3 and the inverted signal of JI7 are added together to form

input channel A. J I 4, J I 5, J I 6 and the inverted signal of J I 8 are added together to form input channel B.

The knobs above the connectors (R1-R8) control the gain (contrast) of each individual input.

The amount of channels A and B mixed into the output, J O 1 through J O 4, is dependent on the position of R9 and the voltage inputted to J I 9. the effect of the knob position and the voltage are additive; the knob to the left and / or a maximun



negative voltage on J I 9 will cause channel B to be outputted only. Similiarly, the knob to the right and / or a maximum positive voltage will cause channel a to be outputted only.

The knob at approximately the center with no voltage applied to J I 9 will cause half of channel A and half of channel b to be added together and outputted-

Dan Sandin

# Function Generator (X3)

The Function Generator generates an output which is an arbitray function ( with up to two points of inflection ) of the input at J 1 1. This results in an effect that is similiar to but more complex and controllable than photographic solarization.

The functions is controlled by R1, R2 and R3. R1 controls the slope of

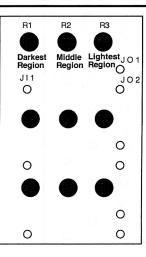
the function for large negative inputs. R2 controls the slope of

R2 controls the slope of the function for inputs near 0 voltages.

R3 controls the slope of the function for inputs of large positve voltage

Clockwise is positive
e. Counter-clockwise is
negative slope.

- Dan Sandin



# .mplitude Classifier

The amplitude Classifier takes an input signal at J19 and seperates it into eight contiguous regions varying from white to black. The value put out by each region is controlled by R1 through R8 and by signals inputted to J11 through J18. The output signals are available for each region seperately at J O 1 through

J O 8.

The sum of these signals is available at J O 9 throgh J O 12. The effect of J I 1, J I 2, R1, R2 etc, is additive in each region.

region.
R9 controls the gain of the signal inputted at J I 9, and R10 generates a bias ( or

constant gray level proportionate to the knob position) which is added to the input signal.

JI1 JO 1 0 0 JI2 JO 2 0  $\circ$ R 9 JI3 О JI 4 JO 4 О J19 JI 5 10 5  $\circ$ 0 JI 6 JO 6 0 JI 7 റ JI 8 0 J O 9- 12

In general, R9 and R10 are used to match the incoming signal to the lightest (top) or darkest (bottom) portions of the 8 regions

- Dan Sandin

# Differentiator ( X 3 )

The differentiator produces an output which is proportionate to the rate of change of the input signal. Fast rates of change correspond to edges in a picture and are preferrentially amplified by the module. J I 6 amplifies only the sharpest edges J I 5 amplifies the sharpest edges and slightly softer edges.

J14, J13, J12 amplify progressively softer and softer edges until by J11 almost all of the picture is amplified.

- Dan Sandin

lly	)	O O	O 0	0
	0	0	0	0
	0	0	0	0
	0	0	0	0

.11.2

.11.3

101

2

JI1

0

# Comparator ( X 3 )

The comparator produces an output which is +.5 volts ( white ) if the input voltage a 1 is greater ( more positive) than the voltage at J12.

The comparator produces an output which is -.5 (black) if the input voltage at J I 1 is less (more negative) than the voltage at J I 2. With 0 volts or no input, the output will either be +.5 or -.5 volts into a 75 ohm load, depending on history.

The variable resistor ( pot ) R1, determines the positive feedback which controls the tendency of the module to stay in the state it is in. Typically it is turned fully clockwise - Dan Sandin

# Oscillator (X2)

The oscillator generates a triangle wave output available at J O 1 and J O 2, and a sine wave output at J O 3 and J O 4 when SW2 is up.

When SW2 is down, the oscillator generates a sawtooth wave at JO 1 and JO 2, and a "s" wave at JO 3 and JO 4.

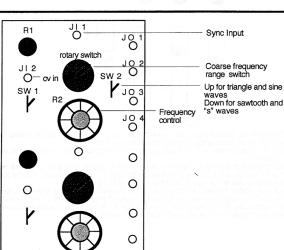
The rotary switch sets the

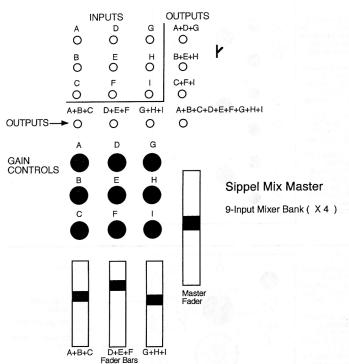
The rotary switch sets the coarse frequency range from 1/100 Hz to 1/2 Mhz. R2 is a continuous frequency adjustment.

If SW1 is up, a signal inputted to J I 2 will control the frequency of the oscillator in combination with R2

When SW1 is down, the veltage control is disabled to the oscillator is more stable.

A sync level (4 volt) signal into J I 1 will trigger the osc to stabilize patterns- D.S.





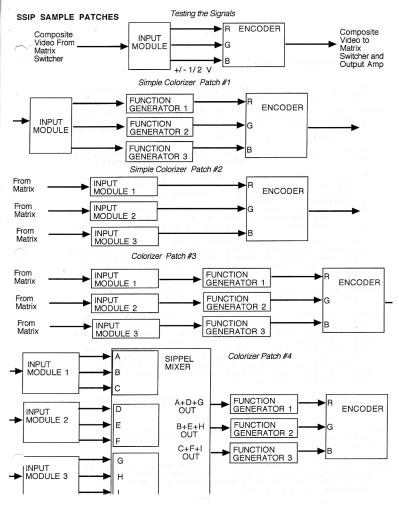
The Sippel Mix Master has (four) 9-input mixer banks. Each bank can be divided into(three) 3-input mixers or grouped together into one 36-input mixer.

The mixer processes +/- 1/2 volt signals and is designed to be interconnected with the Sandin IP modules.

IP modules.
For each set of nine inputs there are seven different outputs. One is the sum of all nine inputs. The other six outputs are the sums of various combinations of inputs. The switch at the top right will group

these inputs over to the next 9-input mixer bank to its right.

One application of these modules is as a final mix before outputting to the encoders. For example, the (A+B+C) output can be sent to the Red input of an encoder, the (D+E+F) output to the Green input and the (G+H+I) to the Blue. Each of the 9-Input Mixer banks can be sent to a seperate encoder with one left over for additional mixing.



#### VIDEO - TECHNICAL GLOSSARY

- CHROMA LEVEL: The color saturation of the video signal, usually measured with a vectorscope and adjustable with a proc amp.
- CHROMA PHASE: A timing function that determines all of the hues in the chrominance signal, represented in degress (0 to 360) relative to color burst, usually measured with a vectorscope, and adjustable with a proc amp.
- CHROMINANCE: Anything pertaining to the color characteristics of the video signal, such as chroma level and chroma phase
- COLOR BURST ( BURST ): A signal placed at the beginning of each horizontal line and used as a reference for the chrominance information. On the waveform monitor, it is represented by a "blip" that extends +20 to -20 ire units and is located to the right of horizontal blanking and to the left of the luminance information. It is necessary for the proper recording of a video signal, even a monochromatic one.
- COMPONENT VIDEO: A method of seperating the luminance and chrominance information in both the transmission and recording of the video signal in order to reduce noise and improve the quality of the image. This capability is specific to certain formats
- COMPOSITE VIDEO: The combined luminance, chrominance, and sync information of a video signal
- EXTERNAL SYNC: The capabalility of some video equipment (e.g. cameras, TBC's, switchers) to be "locked" to another device providing sync, such as a sync generator, in order to be used in a larger system. Sometimes this is done by providing a reference video signal, such as black burst, to the genlock input of the particular video device. At other times it is accomplished by giving specific sync signals from a sync generator (e.g. composite sync, subcarrier, blanking) to a video device as required.
- FIELD: Half of a frame, made up of 262.5 horizontal lines of video information that are scanned in one sixtieth of a second.
- FRAME: One complete scan ( composed of two fields ) from top to bottom of the 525 lines of video information. There are thirty frames in one second.
- GAIN: Also called "video level", it is the amount of contrast in a video signal, measured with a waveform monitor in "ire" units, ( with 100 ire units representing full gain ) and adjustable with a proc amp.

- GENLOCK: A circuit or feature on a particular video device that allows for it to be synchronized with other video devices.
- HORIZONTAL BLANKING: The period of time on each horizontal line when there is no video information. On a monitor, this is the period when the electron beam is shut off so that it can return from the right side of the picture tube back to the left side in order to begin its next line of scanning. On a waveform monitor, it is represented by a drop to -40 ire units on the left side of the waveform.
- H-PHASE: The horizontal timing adjustment of a video signal that results in shifting the raster from left to right
- LUMINANCE: Anything pertaining to the monochromatic characteristics of the video signal, such as gain and pedestal
- PEDESTAL: Also called "set-up", it is the brightness information of the video signal, determining its blackest level, usually set at 7.5 ire units on a waveform monitor, but adjustable with a proc amp.
- PROC AMP( PROCESSING AMPLIFIER): a device or circuit that enables changing some parameters of the video signal ( such as gain, pedestal, chroma phase, and chroma level ).
- SYNC (SYNCHRONIZATION ) GENERATOR: A device that functions as a sort of "master clock" for all of the video devices incorporated into a video system. It outputs several types of synchronization pulses, as is required by individual video machines. The six sync signals are horizontal drive, vertical drive, composite sync, blanking, burst flag, and subcarrier. In addition, sync generators often have "black burst" outputs to besupplied to the genlock input of a device to give it external sync.
- TIME BASE CORRECTOR ( TBC ): is a device which compensates for timing errors on a video tape recorder. It has at least three applications:
  - It corrects some ,but not all of the problems inherent in generational loss when dubbing or editing from one videotape recorder to another.
  - It usually has a proc amp built into it to allow for adjustments of luminance and chrominance information
  - 3. It is necessary for using a videotape recorder ( or any video device that, otherwise, is not externally syncable) in a larger video system for the purpose of combining prerecorded tape(s) with other video sources. Each videotape source requires its own TBC.
- ECTORSCOPE: A display device that measures the chrominance information of a video signal, represented by a color wheel with chroma level along the radius of the circle, and chroma phase

- along its circumference. There are specific marks for video's primary (red, green, and blue) and secondary (magenta, cyan, and vellow) colors on the display.
- VERTICAL BLANKING: The period of time during each field of video when there is no picture information. On a monitor, this is the period when the electron beam shuts off so that it can return from the bottom of the picture to the top in order to begin scanning another field.
- VERTICAL INTERVAL SWITCHING: Any switch or change from one video source to another where the actual time of the change takes place during the vertical blanking interval. This is done to avoid any visible artifact or "glitch" during the switch point
- WAVEFORM: A graphic representation of an electronic signal (e.g. video or audio) with time represented in the horizontal axis and voltage in the vertical. Voltage, in video, corresponds to luminance, and in audio, to amplitude.
- WAVEFORM MONITOR: A display device for measuring the luminance and sync information of a video signal. The display usually shows one horizontal line of video, which takes place in 1/15,750th of a second (30 frames times 525 lines). The combined sync and video signal is one volt. The video waveform is measured in ire units. The blanking level (seperating sync from picture ) is at zero ire units. The picture information is ideally the largest range between 7.5 (pedestal level) and 100 ire units.

RANDOM POSSITIVE ATTENUATORS NEGATIVE SIGNAL RAMP WAVE TRIANGLE & MULTIPLES RAMP WAVE GENERATORS & GENERATORS WAVE WAVESHAPER GENERATOR INVERTERS X 2 SAMPLE & HOLDS GENERATORS + RAMP WAVE GENERATOR ATTENUATORS TRIANGLE WAVE GEN MP WAVE GEN WHITE HOISE 5 TRI OUT PAMP IN RAMPOUT IN OUT WC ATTEN NOISE PULSE OUT TRI OUT PULSE VE ATTEN SUSTAIN PAOUT PAOUT Т SHAPE ATTEN OUT OUT ٥ SINEOUT OUT OUT HOLDIN LED PULSE LED ٧C νc OUT vc ٧Č PULSE MANUAL GATE INVERTER XX ONE MULTIPLE INVERTER XX 0 SWITCH SWITCH OUT QUT IN OUT IN OUT ٧c MULTIPLE TRIANGLE WAVE GEN + RAMP WAVE GENERATOR -RAMP WAVE GEN OUT OUT IN š TREMELO PULSE T OUT MULTIPLE ā RATE ٥ 007 اتنه OUT OUT CAPACITOR, DUAL. VERT ICAL & MIXERS & VOLTAGE-ATTENUATORS VOLTAGE-VOLTAGE-HOR IZ. DRIVE, MULTIPLES CONTROLLABLE CONTROLLABLE CONTROLLABLE INVERTER. FILTER MULTIPLES AMPLIFIER OSCILLATORS COMPARATORS ATTENUATORS MIXER VCO CAPACITOR r,o H.SYNC SYNC OUT 0UTS V. SYNC VERT D. HORIZ D. īN 2 GAIN IN FREG I GAIN INVERTED INVERTER QUTS . 3 OUT IIN 2 IN 114 OUT OUT EXT VC ATTEN. VC ATTEN, SYNC COMPARATOR 

IN REF

NC OUT

REF

COMPARATOR

OUT

IV/OCT

RAMP

TRI SINE MULTIPLE

MULTIPLE

MULTIPLE

114

MIXER

OUTS

INVERTED OUTS O VC

LIMIT

1+2 1+2

IN

I VC

1 2

1+3 1-2

BP HP

VC

ZYC

The Center's Analog Synthesizer is a modular system for creating different types of chancing voltages. The final outputs of the Synthesizer will be used in two distinct ways:

1) as control signals for the image processing modules

as synthesized audio for sound tracks

The system has been designed specifically to produce slowly changing voltages suited to controlling various parameters on the imaging modules such as fading, keying, colorization and so on. One way to view the Analog synthesizer is as a machine that combines various simple changes (represented by changing voltages, which are usually periodic in nature) for the purpose of obtaining interesting and complex changes in imagery. Our model has been the modular electronic music synthesizer, and we have retained many specific standards from that industry. The Analog Synthesizer is compatible with most commercially made audio synthesize equipment, so you are encouraged to bring your own synthesizers to the studio if you wish. If you plan to do so, please check first with the Center's staff to be sure that the appropriate cables are available.

To use this machine you will have to think about voltage control. We can think of voltage as a quantity that can be measured discretely, for example a given number of volts, 5V. For this to be meaningful we need a point of reference. Usually qround (earth) potential is used as a reference as 8V. All sources of constant voltage (DC which stands for direct current) are signed. They have a positive and negative terminal, like a battery. If the negative terminal were connected to ground, then our 5V source will have +5V with respect to ground at its positive terminal. If the positive terminal were connected to ground instead, then we would have -5V at the negative terminal with respect to ground. In the Analog Synthesizer we use both polarities of voltage with respect to ground, and the term "bias" to describe this relationship.

Our control voltage standard is -5V to +5V. Both the Analog Synthesizer and the video image processing modules have been built to respond to this voltage range. This means that a certain aspect of a module's output will be varied from one useful extreme to the other by a sweep of -5V to +5V. This range can be expressed as 18V peak to peak, abbreviated as 18V P-p. For example: in controlling video pedestal, -5V at the control input would produce black while +5V would yield a white pedestal level, with values in between producing various grey levels. You will find that the useful range will lie somewhat within the -5V to sweep the effective range. Since the Analog Synthesizer puts out 18V P-p, it is a simple matter to reduce this voltage to the dessired level.

Most of the control knobs on the video modules have

associated jacks for taking in control voltages. A changing control voltage applied to one of these jacks has the effect of "automatically" turning the knob. Advantages of using a control voltage instead of just turning the knob include (1) variations which are smoother or quicker than would be possible by hand, (2) increased precision and (3) step-like movement. Manual control is useful for complex changes.

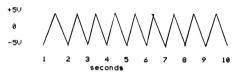
There are three important aspects of a control voltage:

1. GAIN the total voltage excursion, expressed in volts peak to peak

2. BIAS the relative placement of the waveform relative to oround

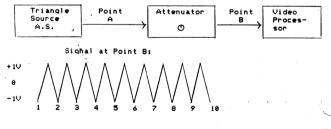
o ground 3. MAVESHAPE the nature of change with respect to time

We represent these aspects graphically in the example below by plotting voltage (expressed along the Y axis) with respect to time (expressed along the X axis).



In this example, the GAIN is 18V P-p, the BIAS is centered at ground or 8V, and the WAVESHAPE is called triangle. The rate or FREQUENCY of this signal is 1 Hertz, expressed as 1 Hz. This means that the waveshape takes one second of time to complete its form and beain to repeat itself. A triangle waveform is called PERIODIC because it repeats itself on a regular basis within a given period of time. Hertz (Hz) refers to the number of cycles the waveform goes through in one second.

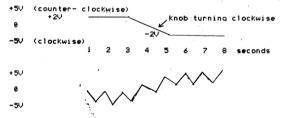
To illustrate how such a control signal can be used in the video synthesizer, let's take this example signal and look at how the patch would be set yp.



The attenuator, which means reducer, is simply a potentiometer (pot) or knob that enables us to lower the gain of a signal with respect to ground. An ordinary volume control is an application of this type of device. In the patch, Point A is the example signal from the Analog Synthesizer. This is patched to the input of the attenuator. Depending on the knob setting, we can obtain any peak to peak value less than 18V P-p at the output of the attenuator. The graph "Signal at Point B" shows the attenuator output, arbitrarily selected to be 2V P-p. This is the signal we will apply to a control input on a video module. Here we must examine how the control knob on the video module interacts with the incoming control signal.

The first point to understand is that the knob is also a source of a control voltage. If there is no input to the VC input jack, then the knob voltage is the only one used. When a voltage is applied to the VC input, it is mixed with the knob voltage. Since it is the nature of our mixers to invert (change the sign) of the incoming voltage, the knobs have been made so that counter-clockwise represents +5V and clockwise represents -5V. This is so that after the mixer, the voltage will be positive for clockwise-of-center rotations. Thus turning the knob clockwise results in black to white, low to high and so on. This also means that, with the knob centered at 8V, a control input of +5V is the same as having the knob all the way counter-clockwise with no control input. You must think of a falling (negative going) control signal as producing the effect of turning the knob clockwise.

Belowis a graph of the voltage output from a knob. The following graph represents an inverted mix of this knob voltage and our attenuated control signal. You can see how the relative placement of the control signal is moved by the knob. This is called BIASING the signal



Let's continue to analyze how the attenuator and bias knobs allow us to get the effect we want. As an example, let's control the clip of a Keyer. With no control voltage input, the extremes that we wish are at these knob settings: 

and 

which are about one-third of the total knob rotation. From this we can

quess that we will need about 3 or 4 volts P-pk to get the entire sweep we require. As long as we start with a larger signal, the attenuator will allow us to achieve the proper gain. With the signal applied to the VC input, the control knob can now be used to bias the signal so that the start point and end point of the effect are where we want them. At this point it is a matter of adjusting the knobs while observing the image in order to set up the desired effect.

Now we will look at the different types of signals in the Analog Synthesizer. A signal can most generally be defined as a voltage that changes with respect to time. In the Analog Synthesizer we can think of a "signal" as the changing voltage of an oscillator that we are going to process and use as an input to video image processing modules. The terms signal and control signal are relative and refer only to use. In general, a control is lower in frequency than the signal it is modifying.

In the Analog Synthesizer signals are classified according to bias, function and waveshape.

#### BIAS

- 1) +- Signal This type of signal usually remains within  $\pm 50$ . However if several of these are added in a mixer, the total possible range is about  $\pm 120$ . This is the maximum voltage excursion in the Analog Synthesizer.
- 2) + Signal This type never goes negative and usually stays in the  $\theta$  to +5V range. Certain modules output this type of signal, and if a negative signal is presented at their inputs it will not pass through. The output will hold at  $\theta V$ .

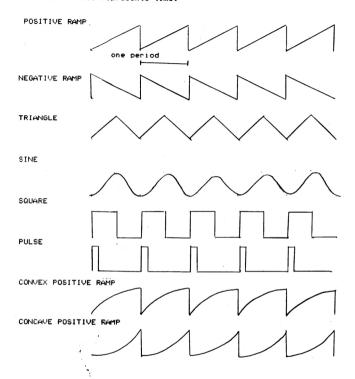
#### FUNCTION

- 1) PULSE This is sometimes called TRIGGER. It is a short Dositive voltage excursion from 0 to +10V and back. It is the point in time that the voltage jumps high that is significant. This point may be used to determine the start of other events in the system.
- 2) GATE A signal which is either 8V or +5V. 8V signifies OFF while +5V signifies ON. This acts like a switch to control the duration of an event, or how long it goes on.

#### WAVESHAPE

Signals classified according to waveshape are named for their graphic representations. See the following page. These periodic waveforms repeat. The number of periods in one second is the frequency in Hertz. The terms positive and negative ramp refer to direction of change, not to bias. Note that inverting (turning upside-down) a ramp or pulse changes the shape, while inverting a triangle, sine or square does not.

These are commonly used waveforms. The YX axis represents voltage. The XY axis represents time.



What happens if the pulse output is patched to the main

events a mount step, when the connection is made the main butger crees quickly following the rising input voltage, but as some as

# DETAILED MODULE DESCRIPTION

The Analog Synthesizer has two boxes each with its own set of modules. The panel controls are listed first and then the functions are described.Module panels have the following features:

1) Signal inputs 2) control inputs 3) one or more outputs 4) knobs for setting initial points (bias knobs) 5) knobs for inverting or attenuating control voltages and 6) special devices such as switches or lights. Inputs are generally to the left and top while outputs are to the right and bottom. Black lines between jacks indicate internal electrical connections.

#### NEGATIVE RAMP

RATE	Bias pot setting initial rate of fall
VC	Attenuator/inverter pot with control voltage input
IN	Main signal input
OUT	Main signal output
P	Pulse Output
LAMP	Output level indicator

The function of this module is to limit the rate at which the output can follow as the input voltage falls. This rate is initially set by the RATE bias control, with more clockwise settings corresponding to higher rates. The rate may also be varied by a control voltage applied at VC. The knob above VC attenuates and/or inverts the control voltage in this manner. At the center of its rotation, an incoming control voltage will be most greatly attenuated, that is will have the least effect. Turning the knob to the right, clockwise, will cause a positive voltage being presented at the VC input to effect a higher rate. Counter-clockwise rotation with a positive VC input will lower the rate. The farther the knob is turned in either direction, the greater the effect. Negative voltages may be applied to the VC input.

The negative ramp module puts out only positive voltages, thus the input must be positive to pass through. There are two outputs. The main signal output will accurately follow a rising voltage at the input, while limiting the rate of a falling input. The pulse output remains at +180 while the main output is close to 8V. If the main output rises above about .2V, the pulse output will go 8V. The lamp indicates main output voltage, getting brighter as the main output increases in voltage. If the output frequency is rapid, the light will appear to remain constant at a medium brightness.

What happens if the pulse output is patched to the main signal input? First, the output has been at 0, so the pulse output is high, +10V. When the connection is made the main output rises quickly following the rising input voltage, but as soon as

the main output rises this causes the pulse output to go to  $\theta V$ . Since the main output can't fall any faster than the control will allow, we will get a falling ramp at the main output. When the ramp approaches  $\theta V$ , the pulse output jumps high and the process is repeated. Thus we have an oscillator with ramp and pulse outputs.

When patched as above, the two negative ramp modules on the Analog Synthesizer each have different frequency ranges. These are shown below.

MODULE	R	ANGE		w	i th	CONTR	ROL	VOLT	AGE
Top Negative Ramp Bottom Neg. Ramp Top Positive Ramp Bottom Pos. Ramp Top Triangle Bottom Triangle	17 23 7 120	sec/cycle sec/cycle sec/cycle sec/cycle sec/cycle sec/cycle	to to to	1000 1000 2000 1700	Hz Hz Hz Hz	up up	to to	1200 2500 1700 4000	Hz Hz

# POSITIVE RAMP

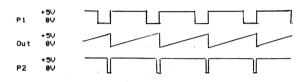
RATE	Bias pot setting initial rate of rise
VC	Attenuator/invertor pot with control voltage input
IN	Main signal input
OUT	Main signal output (TOP)
P1	Pulse output one
P2	Pulse output two
START	Input for starting ramp
SUS	Input for starting ramp and sustaining at +50
Switch	Performs gate function for sustain
VC	Voltage control of waveshape from lower output
OUT	Sine type waveform output (BOTTOM)

The function of this module is to limit the rate at which the output can folllow as the input voltage rises. This rate is controlled in a manner similar to the negative ramp. There are four outputs. The main signal output limits the rate of a rising input, while accurately following a falling input. Pli is low while the ramp is above 4.5V and goes high at the point the ramp wave falls. There must be an input to Start or Sus to get an output from P2. P2 stays high while the ramp is rising and holds at 8V if the main output is also at 8V. For both pulse outputs high equals +5V. Like the negative ramp, an oscillator can be patched by connecting P1 to IN. The resulting ramp wave at the main output is about 4V P-p from +.5V to +4.5V. A sine-type waveform is simultaneously available at the lower output, and its shape is voltage controllable. This output is also positively biased and has a gain of about 2V P-p.

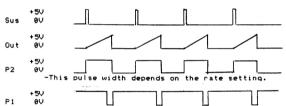
. An oscillator can also be created by patching PI to Start or Sus. This will give an output at P2 as well as raising the main signal output gain to 5V P-p.

A rising ramp may be started by supplying a trigger pulse to the Start input. When this ramp is completed and returns to 8V, it will remain at 8V until another trigger is received. A trigger applied to Sustain will also start a ramp. If a gate is applied to Sustain, a ramp will be initiated also, and if the gate is still present when the ramp has reached its maximum, the main output will hold at +5.5V until the gate is removed. A gate patched to Start will perform the starting function, but will not sustain the output.

#### Pl Patched to Sustain



# External Pulse Patched to Sustain



# WAVESHAPER (WS)

OUT

GAIN	Input, attenuator
SHAPE	Bias pot setting initial waveshap
VC	Control voltage input #1
VC	Control voltage input #2
IN	Main signal input

Main signal output

The purpose of this module is normally to convert a positive ramp wave into a sine-type wave with a voltque controllable waveshape. For proper operation in this mode the input should be a positively biased positive ramp wave (such as the output of a positive ramp) with a qain of about 2V P-p. An input attenuator is provided for setting this gain.

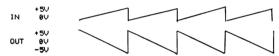
The shape control is a bias pot which sets the initial waveshape. Centered, the output is an approximate sine wave. Varyiong the control results in a more ramp-like waveform at the output. The extremes of rotation produce nearly concave positive or negative ramps. The output is positively biased with a gain of about 5V P-p.

The VC input directly below the shape control allows voltage control similar to the shape control pot. With the pot centered, +-5V is sufficient to sweep the pot range. The lower VC input has the effect of biasing the output so as to cut off the lower portion of the signal, which also changes the waveshape.

Other waveforms may be applied to the input. Note that at audio frequencies the WS is used as a timbre modifier.

#### INVERTER X2

This module ,which has no controls, is designed to convert a 0 to  $\pm 50$  signal to a  $\pm 50$  signal to Since it also inverts, the in/out relationship looks like this:



You can see that an input of 8V (no input) will leave the output at  $\pm 5V$ .  $\pm 2.5V$  in comes out 8V.  $\pm 5V$  in comes out 8V.  $\pm 5V$  in comes out 5V. The gain is doubled. A hint: If you want to invert but leave the output still at 8V to  $\pm 5V$ , simply attenuate the input to 2.5V. Note that a negatively biased input will cause the output to go above  $\pm 5V$ . This module is meant to increase the gain available from the Ramp modules. Since the output from the Triangle patched to oscillate is  $\pm 1V$  to  $\pm 4V$ , passing this signal through Invert X2 will give  $\pm 7V$  to  $\pm 3V$  out. This is still within a usable range.

#### RANDOM

- WHITE 1 White noise output with energy concentration in center
- WHITE 2 White noise output with energy concentration at
- PINK Pink noise output filtered from WHITE 1
- RAMP + Fast random ramp output, positive bias, 5V P-p
- RAMP +- Fast random output, +-bias, 10 V P-p
- SAMPLE Button changes Step outputs to new random level
- SAMPLE Jack accepts triqger pulse input for above function
- STEP + Random held voltage output, + bias
- STEP +- Random held voltage output, +- bias
- TREM 7 Hz sinewave output with randon amplitude

Q Varied randomness is TREM output
PULSE Random pulse output

RATE Varies range of random pulse frequency

The random module is a source of different kinds of randomly changing voltages. Random sources are categorized by their average rates of change. White noise contains the widest range of frequency variations: the entire audio range. 20 Hz to 20 KHz. Pink noise contains less high frequencies than white noise, while the random ramps are low in frequency. Patching these sources to an audio monitor will give you a good idea of this. The step outputs have a rate of change equal to the sample pulse rate.

White 1 output is biased about ground and is generally 5V P-p. The probability for an instantaneous output is higher for values nearer to ground. White 2 is 8V P-p biased at ground, with greater probability of an instantaneous output being at either + or -4V.

The Step outputs are from a sample- and- hold- connected internally to a random source. Pushing the button or supplying a pulse to Sample (the only input on the module) will set the Step outputs to new random levels. These voltages drift slightly.

Patching the random tremelo output to an oscillator VC input will guickly show how this functions. Notice a time lag after turning the Q knob before the effect changes.

The Rate control adjusts the random pulse output from a pulse every few seconds to dozens per second. When setting this control allows 15 seconds or so to observe the effect at one setting.

# ONE SHOT

S Sample Input
T Track input
OUT One shot output

This module has no controls. Normally the output sits at +10V. If a triquer pulse is supplied to the S input, the output will fall to 8V for a preset time interval of about 1 ms, then rise quickly back to +10V. A gate to the T input will cause the output to hold at 8V as long as the gate is present.

Patching the output of this module to the Hold input of the Triangle module creates a sample and hold. During this patch, a gate to the Track input returns the Triangle to its following mode.

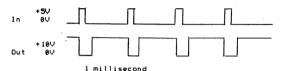
#### DNE PULSE

This module puts out one 100 uS pulse each time the button is pushed. Use this for a manual trigger.

#### GATE

This switch when up applies +6V to its output jack and 6V (ground) when down. Use this for a manual gate (patch to Hold, Track, Sustain, etc.).

# In/Out Waveforms for One Shot



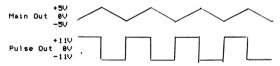
#### TRIANGLE

Lamo

RATE	Bias pot setting initial triangle period
VC	Attenuator pot with control voltage input
IN	Main Signal input
OUT	Main signal output
P	Pulse output (+-11V, 22V P-p)
н	Hold input

Output level indicator

The triangle module limits the rate at which the output can follow both rising and falling voltages at the input. Unlike the Ramp modules, the 'Triangle inputs and outputs may be biased positive or negative. The main and pulse outputs are related in this manner: as the main output falls below -IV, the pulse output goes to +IIV and holds there until the main output rises above +4V. The pulse output then falls and holds at -IIV until such time as the main output again falls under -IV. Patching the pulse output to IN yields an oscillator with the following waveforms:



Another difference from the Ramp modules is that the VC attenuator pot does not invert an incoming control voltage: positive voltages always increase the rate.

The Hold input accepts a +5V or higher gate signal. When the gate is presented the main output will hold at the voltage it had at the moment the gate went high. This same function amy also be accomplished by bringing the VC input to a sufficiently negative value, depending on the initial rate setting.

The lamp is an output level indicator which functions similarly to the lamp on the negative ramp module.

# ANALOG SYSTEM: MODULES IN BOX 2

# VOLTAGE CONTROLLED OSCILLATOR (VCO)

EREO Bias not setting initial frequency, wide range FINE Bias pot setting initial fequency, narrow range VC. Most sensitive control input, with attenuator LIN Least sensitive control input, inverted IUZ0CT

Control input set for 1 V increase to double

frequency HIZLO.

Audio/subaudio range selector H/EXT/U Sync selector switch

FXT Sync innut

RAMP Sawtooth waveform output TRI Triangle waveform output

SIN Sine waveform output

This module is a music quality oscillator. Its function is to generate periodicc waveforms of three shapes, the controlled parameter being the frequency of the outputs. There are a number of control inputs which may be used simultaneously. The outputs are all the same frequency and are all similarly affected by the controls. A HI/LO switch determines the mode of operation. HI mode is that of an audio range oscillator. LO mode is that of a very low frequency control oscillator.

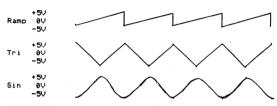
In LO range the frequency is manually variable by the bias pots from 1 cycle every 20 seconds to about 15 Hz. In HI the manual range is 16 Hz to 16 KHz. Both of these ranges can be expanded with voltage control. Negative control voltage will lower the LO range infinitely until oscillation stops. Positive control voltage can extend the HI range up to 100 KHz, however at this frequency the outputs are degraded in amplitude and shape.

All FM (Frequency Modulation) inputs allow exponential control of the frequency. This means that for an increase of a set amount of voltage (for example, 1V) the frequency will be multiplied by some factor. In the case of the V per octave inputs. a 1V increase will double the frequency. The VC input with attenuator allows adjustment of this factor, and at high settings the frequency may be multiplied by a factor of 4 or more for each volt increase of control voltage. The LIN control input is much less sensitive, and operates inversely: an increase of control volgage lowers the frequency. This ratio is set at approximately 6U/octave.

Sync options are selected with the sync switch. Center is

off, while H/EXT normally supplies horizontal sync for use at high frequencies (above 15 KHz). Should a signal be patched to the EXT sync input, this input is selected instead of horizontal sync. Signals patched to EXT should have a sharply rising edge for proper syncing. The V position selects vertical sync. Standard video sync must be plugged to the BNC inputs at the left of the panel for these functions to be used.

All outputs are standard +-5V signals. The phase (relative changes) of the outputs are shown below.



# COMPARATORS

Under the heading of Comparators are three small modules above the Comparator modules. These will be described first.

- 1) <u>Capacitor</u> A patchable capacitor is available between the two jacks containing the symbol for capacitance ( -11- ). Either jack may be the input. Its function is to block the offset bias of an audio frequency signal, so that the output is rebiased around ground.
- 2) U and B Vertical and horizontal sync pulses are available at these jacks, pulses coing from zero to +8V for system compatibility. Example: Start a ramp with a V pulse.
- 3) Invert A standard analog inverter is made available for changing the sign of the input voltage.

#### Comparator Panel

UC.

IN Comparator input
OUT Comparator output

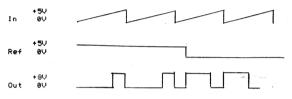
Control voltage input for reference

REF Bias pot for setting initial reference point.

The function of the comparator is to compare two voltages and signal when one is higher than the other. The output is high (+8V) if the input is higher than the reference voltage, and zero volts if the input voltage is lower than the reference. Thus if a periodic voltage is applied to the input and regularly crosses the reference point, a pulse wave is the resulting output.

Since the duty cycle of the pulse wave is determined by the relative setting of the reference, this parameter is variable and may be voltage controlled. This is called pulse width modulation. The comparator may be used to obtain pulse or square waves from any oscillating module. Notice that a square wave is a special case of a pulse wave for which the duty cycle is 50%. At audio frequencies bulse width modulation produced interesting timbral changes in the sound. One useful sub-audio application would be to control a VCA. The VCA would switch rapidly from zero to full gain. An attenuator on the VCA VC input would be helpful.

Below are the comparator input. reference and output as an example to demonstrate the function:



#### MIXERS

1.2.3 Inputs with associated attenuators

"Wild" inputs

- + Bias pots for setting output offset voltage OUT

Main output, non-inverting INVERT Main output, inverting

The mixers enable a combination of the outputs of two or more modules, with manual control of the mix. All four inputs are added together. This means that if you put the same signal into two of the inputs turned up full, the output will be twice the input. The Wild input simply has no attenuator. This allows the mixers to be patched together for a six-input mixer. Any signals in the Analog Synthesizer may be mixed, but note that up to +-11V may result. The bias pot allows offsetting which is useful when mixing control volaces.

#### VOLTAGE CONTROLLED AMPLIETERS (DUAL VCA)

VC >	Control input for panning and fading
PAN	Bias pot setting initial pan/fade
1,2 GAIN	Bias pots setting individual initial gains
1,2 IN	Respective inputs for VCA 1 and VCA 2
1+2 IN	Input simultaneously to UCA 1 and UCA 2

1.2 UC Respective gain control inputs for UCA 1 and UCA 2 1+2 VC Simultaneous gain control input for both UCAs

1.2 OUT

Respective VCA outputs

1+2 Sum of the outputs of both VCAs 1-2 Sum of VCA 1 output plus the inversion of VCA 2

The VCA porvides a means of controlling the gain of a signal by a control voltage. Any signal, including subaudio, may be processed. This module consists of two independent VCAs with provisions for tandem use in various ways.

Each VCA has: 1) an input for the signal to be gain controlled 2) a control input 3) an output. A bias pot sets the initial gain point, then the gain is increased by positive control voltages and decreased by negative control voltages. With the bias pot set for zero gain, a +5V control signal will give unity gain, that is, the output will be indentical to the input. Control voltages up to +10V will produce gains of up to two times the input.

Remember that the bias pot adds to the incoming control voltage so that signal can be doubled by the pot being all the way in combination with a +5V control input. For instance, if a +-5V control were used in this case, the output would be varying from zero to two times the input signal amplitude.

The two VCAs are linked in the following ways. First, there is a third control input at the top of the panel for VC panning and fading. This input affects the two VCAs in opposite fashion: as the control voltage increases, VCA 1 gain increases, while VCA 2 gain decreases. As the bias pot is turned clockwise, the same is true. There is also a fourth control input labeled 1+2 VC. This control input affects both VCAs simultaneously in the same fashion as the individual VC inputs. All inputs may be used at the same time on this module.

There is also a 1+2 input which sends the signal to be modified to both VCA inputs. This is used during panning. For fading and voltage controlled mixing there are two mixed outputs of the two VCAs. 1+2 is a normal additive mix. while 1-2 is a different mix. Note that with a signal patched to 1+2 ONN, the 1+2 output may be as high as 4 times gain, while the 1-2 output can never be higher than 2 times gain.

When using this module, first center all three pots. This sets the initial gain of both VCAs to .5. Turning the pan pot will cause one VCA to go to unity gain as the other goes to zero gain. The individual bias pots can be used to balance the overall effect. Leave the pan pot centered for individual use of the VCAs.

For fading, the two signals to be cross-faded are patched into inputs 1 and 2 respectively. The output is taken from 1+2 or 1-2 output. For panning, the input goes to 1+2 IN and the outputs are taken from the respective 1 and 2 outputs.

VOLTAGE CONTROLLED FILTES (VCF)

HI/LO Audio/subaudio range selector
GAIN Attenuator for Main input

Q Resonance control (also affects gain of Limit

input)
VC Control input with attenuator/inverter for VCQ

FREQ Bias pot setting initial filter cutoff frequency VC Control input with attenuator/inverter for VC

frequency

LIMIT Signal input limited by Q setting

MAIN Main signal input HP High pass output BP Band pass output LP Low pass output

The function of the VCF is to alter the gain of an input signal with respect to that signal's frequency. The filter cutoff frequency is the reference. For the high pass function, input frequencies below the cutoff frequency will be attenuated. The low pass function attenuates frequencies above the cutoff, while the bandpass function attenuates both above and below the cutoff frequency, allowing only frequencies near the cutoff to pass. The Q function determines the sharpness, or closeness to the cutoff frequency, of the attenuating effect.

Since many waveforms contain a complex distribution of different frequencies which determine the timbre of color of a sound, a VCF is often used to modify timbre. Patch a sawtooth wave into the VCF input and manually vary the cutoff frequency. You will notice the common "wah" effect. Listen to each of the outputs and notice the different effects. Notice the action of the Q control. Patch the signal into the Limit input and notice how the signal is attenuated at high Q settings. The purpose of this limiting function is to obtain a more constant output level since the filter "peaks", increases its output amplitude sharply at the cutoff frequency. Higher Q settings produce higher peaks, so the limit function is used to counteract this effect.

Both the cutoff frequency and the Q are voltage controllable, and an attenuator/inverted similar to that on the ramp modules is provided for processing the control voltage coming in.

In LO range, the VCF can be used to modify control volatges. The outputs have a characteristic sinewave pattern, with the BP and LP outputs being 98 decrees out of phase. The BP output leads the LP output by 1/4 cycle. This is different from 1/2 cycle difference produced by a simple inversion. Try patching in a slow ramp and control something with the output.

Finally, the VCF can be made into a VCO by patching the BP output into the Main input. The output will be a sinewave of about +-10V. The Q and Gain controls must be set fairly high for this to work. Notice that if one of these is decreased slowly, the oscillation dies out gradually, which is sometimes useful.

In this oscillator mode, the HI/LO switch can select audio or subaudio oscillations,

# SUGGESTED PATCHES

- Output of an oscillator to its own UC input. This will yield the concave and convex waveforms discrammed earlier.
- 2. Main out of a + ramp to the main input of a ramp. Using the switch on the + ramp you can initiate a rising ramp (taken from the ramp main output) that will hold as long as the switch is up and fall when the switch is down. The rise and fall rates can be set individually.
- 3. Manual gate to the main input of a triangle module. When the switch goes up the main output will rise at the adjustable rate and hold until the switch is turned down, then fall at the same rate it rose (unless you change the rate control in the meantime)
- Random pulse output to the sample input of the random module.
   This will yield random levels at the step outputs at random intervals.
- 5. Pulse output from a ramp (patched as an oscillator) to the sample input on the random module. This will yield random voltage levels at the step outputs at requiar intervals
  - -- Richard Brewster 1978
  - -- revised 1984